



# Cisco TelePresence MCU 4200 Series MSE 8420 Media blade

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## Logging into the web interface

The Cisco TelePresence MCU web interface is used for administering the MCU device, managing conferences, users, and pre-defined endpoints. You can also perform many conference-related tasks using the web interface that you cannot otherwise do.

When connecting to the Cisco TelePresence MCU web interface, you must log in so that the MCU can associate the session with your configured user and a set of access privileges. The MCU has a set of configured users, and each user has a username and password that are used for logging in.

1. Using a web browser, enter the host name or IP address of the MCU.
2. To log in, click **Log in** and enter your assigned **Username** and **Password**.
3. Click **OK**

The main menu appears, restricting the available options based on your access privileges. Administrators have full access; standard users can create new conferences and manage their profiles; guest users typically can access publicly available conferences.

The **Login** page of the MCU displays a welcome banner which administrators can configure to display text relevant to your organization. For more information, refer to [Customizing the user interface](#).

## Failing to log into the web interface

When connecting to the Cisco TelePresence MCU web interface, you must log in so that the MCU can associate the session with your configured user and a set of access privileges. The MCU has a set of configured users, and each user has an ID and password that are used for logging in.

If you see the **Access denied** page, you have not been able to log in for one of the following reasons:

- **Invalid username/password:** you have typed the incorrect username and/or password. If Advanced account security mode is enabled and you incorrectly type the username and/or password three times and if this is an admin account, it is disabled for 30 minutes; for any other account, it is disabled indefinitely (or until you, the administrator, re-enable the account from the **User** page)
- **No free sessions:** the maximum number of sessions allowed simultaneously on the MCU has been exceeded
- **Your IP address does not match that of the browser cookie you supplied:** try deleting your cookies and log in again
- **You do not have access rights to view this page:** you do not have the access rights necessary to view the page that you attempted to see
- **Page expired:** the **Change password** page can expire if the MCU is not entirely happy that the user who requested to change password, is actually the user submitting the change password request. (This may happen if you use a new browser tab to submit the request.)

### Invalid passwords

If Advanced account security mode has been enabled, the MCU will disable a user's account if that user incorrectly enters a password three times consecutively. If this is an admin account, it is disabled for 30 minutes; for any other account, it is disabled indefinitely (or until you, the administrator, re-enable the account from the **Users** page)

# Calling into conferences

Depending on how your system administrator has configured the MCU and conferences on it, you might be able to join conferences by simply dialing a phone number.

## Dialing in using a video endpoint

Your system administrator may have configured the MCU to allow you to use your video endpoint to directly dial the conference by dialing a phone number. You will be required to enter the conference ID, and PIN if required. Or you may be able to dial by IP address and connect to the auto attendant. See [Using an auto attendant](#) for details.

Note that some video endpoints require that you activate the keypad before dialing. For example, you might need to press the # key.

## Dialing in using an audio-only phone

If your phone system allows calls to the MCU, you may use your regular phone to join conferences as an audio-only contributor. You must enter the conference ID (and PIN, if required).

If your regular phone number is linked to your video endpoint, then when you use your phone to join a conference, the video portion of the conference will automatically appear on your video endpoint's screen. You can continue to use your regular phone for the audio portion of the conference. To do this, the video endpoint must be configured as that user's **associated video endpoint**. For more information, refer to [Adding and updating users](#).

The advantages to this method are that you are able to use the same method and phone to make video calls as you do traditional audio calls. You also may have improved audio signaling. However, this method requires significant configuration and setup from the system administrator. Your company's network may not have all the components available to support this method.

When in a conference using an audio-only phone, pressing \*6 mutes your phone such that you will not be contributing audio to the conference; pressing \*6 again unmutes your phone.

## Using an IP VCR recording as the video contribution

Audio-only conference participants can show a recording from a Cisco TelePresence IP VCR as their video contribution. To do this:

Either register the MCU and the IP VCR with the gatekeeper or add the IP VCR as an H.323 gateway.

1. Go to **Endpoints > Add H.323 endpoint** and add the recording as an H.323 endpoint:
  - i. Set the address as the recording number.
  - ii. If you are using a gatekeeper, leave the other fields at their default settings; if the IP VCR is set up as an H.323 gateway, select the VCR from the **H.323 gateway** list and leave all other settings as default.
2. Go to **Users > Add user** and set up the user using [Adding and updating users](#) for more information and setting the **associated video endpoint** as the IP VCR recording.

## Dialing in using ConferenceMe

The only method for dialing into a conference directly from ConferenceMe is by entering a HTTP link that has been emailed to you from the conference administrator.

For more information on joining a conference using ConferenceMe, see [Using ConferenceMe to join conferences](#).

# Using an auto attendant

Your system administrator may have set up an auto attendant for you to use to join conferences. An auto attendant presents you with a series of menus from which you can choose a conference to join.

For further information about accessing conferences (including the use of DTMF tones), refer to the document "Accessing Conferences", available in the Documentation area of the web site.

## Calling an auto attendant

There are typically two ways to call an auto attendant using your video endpoint. Your system administrator should provide you with information about which method you should use:

- ▶ Enter the IP address or host name of the MCU device
- ▶ Dial using a standard E.164 phone number

## Accessing the main menu

When you successfully connect to the MCU, the main menu displays on your video screen, and you should also hear the audio instructions.

Navigate the auto attendant using the Far-End Camera Controls (FECC) on your video endpoint. Use the up and down controls to highlight the option or item you require; use right to make your selection. To return to a previous menu from a sub-menu, use left.

You can jump to the end of the menu when at the start by using the up control; similarly, you will loop back to the start if you are at the end and use the down control. Note that there is a scroll bar in the bottom right of the video display to indicate where you are in the auto attendant menu. From anywhere in the menu, you can jump to the first entry with #2 and to the last entry with #8.

By default, you join a conference by using the number keys on your endpoint to dial the conference's numeric ID, followed by a '#'. If a conference has both a **Numeric ID** and a **Guest numeric ID** set, you may enter either ID, and will join the conference as either a chairperson or guest as appropriate. As you start to enter a numeric ID, the sequence you have typed will be shown at the base of the auto attendant screen. You may cancel the numeric ID entry (for instance to correct an error) by pressing '\*#'.  
\*#.

If you have connected to the auto attendant using an endpoint that has no FECC capability (for example many SIP endpoints), you can use the number keys on your endpoint to navigate the menus; this is called "DTMF navigation mode". DTMF navigation mode enables you to use the number keys: 2, 4, 6, and 8 in the place of up, left, right, and down respectively.

To enter DTMF navigation mode:

- ▶ on connecting to the auto attendant, press the pound (hash) key twice as follows:  
##  
The message "DTMF menu navigation enabled" appears briefly at the bottom of the auto attendant display.

To exit DTMF navigation mode:

- ▶ press the pound (hash) key twice as follows:  
##

The message "DTMF menu navigation disabled" appears briefly at the bottom of the auto attendant display.

When in DTMF navigation mode, you will not be able to use the number keys to enter the numeric ID of a conference.

Typically, you will have these options:

### **Create new conference**

This option enables you to start a new conference that takes place immediately (an "ad hoc" conference). When creating this conference, you'll need to add a conference ID and you might also be required to set a PIN (to restrict access to the conference). Other participants are then free to join the conference as they would any other, for example by using the auto attendant, or by calling in directly (see [Calling into conferences](#)).

As the system administrator, you can disable this option so it is not available on an auto attendant.

### **Join a conference**

All permanent conferences or conferences scheduled to take place (at this time) can display here, listed by the conference name. Your system administrator can limit the conferences that display here. So, if you do not see the conference you want to join, verify the conference name, start time, and finally verify that your system administrator has enabled the conference here.

### **Access other auto attendant menus**

Your system administrator may allow this auto attendant to access additional auto attendants, giving you access to additional conferences.

## **Participating in the conference**

After you join a conference, the in-conference controls and features are the same regardless of how you joined the conference (see [Using in-conference features](#)).

## **Leaving a conference**

When you want to leave a conference, you can either simply hang up, or if you joined the conference from an auto attendant and want to return to the auto attendant:

- if you are using FECC, press zoom out. The zoom out menu is displayed; choose **Return to auto attendant**
- if you are using DTMF tones, press 7. The zoom out menu is displayed. Press 8 to navigate down through the options and 6 to select **Return to auto attendant**

## Using streaming to view conferences

The streaming interface enables you to enter the ID of a conference and either view it in a web browser on your PC or join the conference using the ConferenceMe software endpoint on your PC. You do not have to have a user account on the MCU to be able to do this.

By default, the streaming interface is accessible by all users (even those who have not logged into the MCU). However, administrators can disable public access to the streaming interface. To do so, go to **Settings > User interface**.

If you do need to log in to the MCU, use the **Log in** link on the top right of the screen.

### Stream a conference

Refer to this table for assistance streaming a conference. After you have completed the fields, click **Stream this conference**.

Field	Field description	More information
<b>Sign-in name</b>	A name by which you will be identified should you use the text chat facility.	Note that text chat will not be available unless the MCU has the Web conferencing feature key
<b>Conference ID</b>	The numeric ID that uniquely identifies the conference you want to stream.	
<b>PIN</b>	If the conference that you want to stream is protected by a PIN, enter it here.	
<b>Streaming rate</b>	Select the streaming rate you prefer to use.	There are usually two configured options for streaming rate from which you can select.

### Advanced streaming options

You can enter advanced streaming options by clicking the **show advanced streaming options** link.

Field	Field description	More information
<b>Prefer multicast</b>	When enabled, multicast will be used when streaming this conference.	<p>This option is not available if the streaming rate specifies that Windows Media Player will be used.</p> <p>When enabled, the MCU's streaming page will attempt to access the multicast meadi stream for the chosen conference. Note that if the streaming page fails to stream the conference, you can return to the this page and clear this option; doing so will cause the MCU to attempt to stream the conference using unicast.</p>
<b>Play audio and main video</b>	Select this option to stream the audio and video.	If you clear this option, then neither the audio nor video will play.

Field	Field description	More information
<b>Video size</b>	Select a size (resolution) for the video.	This option affects the size of the media player when it opens.
<b>View content channel</b>	Select this option to stream the content channel.	<p>If you clear this option, the content channel will not play. Clearing this option also disables text chat.</p> <p>Note that text chat will not be available unless the MCU has the Web conferencing feature key; and that without the Web conferencing feature key, this option is grayed out.</p>
<b>Content size</b>	Select a size for the content channel.	

## Join a conference using ConferenceMe

To join a conference using ConferenceMe, enter the numeric ID of the conference you are joining in the **Conference ID** box and click **Join conference**.

This option is only visible if you have the Web Conferencing feature key installed.

## Using streaming to view conferences

The Cisco TelePresence MCU supports streaming, which involves sending conference media (audio and video) to a remote computer, allowing the user to watch and listen to a conference. The media flows in just one direction, so it is not possible to actually contribute to a conference via streaming.

The streaming video may be delayed by approximately 4-10 seconds compared to the real-time video.

Streaming must be enabled on your MCU before you can watch a conference. To enable streaming, go to **Settings > Streaming**. See [Configuring streaming and ConferenceMe settings](#) for more information.

To view an active conference using streaming:

1. Go to **Conferences** and click **Stream** next to the conference name.
2. Select the Streaming rate. Refer to the table below for tips.
3. If you want to display and select advanced streaming settings, click the link. Select the settings you require, referring to the table below for tips.
4. Click **Start streaming** to display the conference.

Field	Field description	More information
<b>Streaming rate</b>	The configured options for streaming video.	The two options available here have been configured on the <b>Settings &gt; Streaming</b> page.
<b>Prefer multicast</b>	If multicast is enabled for the selected streaming rate on the <b>Settings &gt; Streaming</b> page and you select <b>Prefer multicast</b> , then multicast will be used when streaming this conference.	Multicast streaming allows an unlimited number of viewers to watch a conference on a MCU. Note that you cannot use Windows Media Player to stream in multicast mode on the MCU.
<b>Play audio and main video</b>	Select this option to stream the audio and video.	If you clear this option, then neither the audio nor video will play.
<b>Video size</b>	Select a size (resolution) for the video.	This option affects the size of the media player when it opens.
<b>View content channel</b>	Select this option to stream the content channel.	If you clear this option, the content channel will not play. Clearing this option also disables text chat. Note that text chat will not be available unless you have the Web conferencing feature key; and that without the Web conferencing feature key, this option is grayed out.
<b>Content size</b>	Select a size (resolution) for the content channel.	This option affects the size of the media player when it opens.

# Using ConferenceMe to join conferences

The Cisco TelePresence MCU allows users without access to a hardware-based endpoint to join conferences using the ConferenceMe software endpoint if it is installed on a participant's computer.

ConferenceMe allows participants to access a conference without using a hardware-based endpoint. ConferenceMe works with Cisco TelePresence MCU units and MSE media blades. Using ConferenceMe, there are two methods for entering conferences: through the web interface of your MCU, or by launching the program and entering the conference address.

Before your MCU is able to accept calls from ConferenceMe, a feature key must be installed on your MCU. ConferenceMe is included in the Web conferencing option. If you have purchased the Web conferencing option for your MCU, you do not need to install an additional feature key. For more information about configuring the MCU to use ConferenceMe, refer to [Configuring streaming and ConferenceMe settings](#).

From within the MCU interface there are three methods for join a conference using ConferenceMe:

- ▶ [Through the Conference List page](#)
- ▶ [Through the Participant List page](#)
- ▶ [Through the MCU Streaming page](#)

## To join a conference using the conference list page

1. Go to **Conferences** and click *ConferenceMe* next to the conference name.
  - If you already have ConferenceMe installed, select the **Click here to join the conference using ConferenceMe** link.
  - If you do not currently have ConferenceMe installed:
    1. Select **Click here to install**.
    2. Complete the installation process by following the on-screen instructions.
    3. When the installation is complete select **Click here to join the conference using ConferenceMe**.
2. If you want ConferenceMe to open automatically and connect you to your conference on your next visit, select **Automatically join next time**.
3. If the conference is PIN-protected, type the PIN using the number keys on your keyboard.
4. ConferenceMe opens and you are connected to your conference.

## To join a conference using the participant list page

1. Go to **Conferences** and click the conference name.
2. Click **Join**.
  - If you already have ConferenceMe installed, select the **Click here to join the conference using ConferenceMe** link.
  - If you do not currently have ConferenceMe installed:
    1. Select **click here to install**.
    2. Complete the installation process by following the on-screen instructions.
    3. When the installation is complete select **Click here to join the conference using ConferenceMe**.
3. If you want ConferenceMe to open automatically and connect you to your conference on your next visit, select **Automatically join next time**.
4. If the conference is PIN-protected, type the PIN using the number keys on your keyboard.
5. ConferenceMe opens and you are connected to your conference.

## To join a conference using the MCU Streaming page

1. Navigate to the **Streaming** page of your MCU. You will see a section called Join a conference using ConferenceMe.
2. Enter the Numeric ID of the conference you are joining in the **Conference ID** dialog box.
  - If you already have ConferenceMe installed, select the **Click here to join the conference using ConferenceMe** link.
  - If you do not currently have ConferenceMe installed:
    1. Select **click here to install**.
    2. Complete the installation process by following the on-screen instructions.
    3. When the installation is complete select **Click here to join the conference using ConferenceMe**.
3. If you want ConferenceMe to open automatically and connect you to your conference on your next visit, select the **Automatically join next time**.
4. If the conference is PIN-protected, type the PIN using the number keys on your keyboard.
5. ConferenceMe opens and you are connected to your conference.

# Using in-conference features with video endpoints

After you join a conference, you can control many conference features directly from your video endpoint. (For information about in-conference features using the web interface, see [Controlling in-conference features](#).)

- ▶ [Controlling conference views](#)
- ▶ [Understanding participants status](#)
- ▶ [Using far-end camera controls](#)
- ▶ [Understanding in-conference status icons](#)
- ▶ [The zoom out menu](#)
- ▶ [The conference welcome message](#)

For further information about accessing conferences (including the use of DTMF tones), refer to the document "Accessing Conferences", available in the Documentation area of the web site.

## Controlling conference views

Your video endpoint typically has navigation keys, such as up/down and left/right (on a keyboard or remote control), that allow you to control the camera viewing angles. When connected to a conference, you can also use these controls to scroll through participants and conference layout formats (see [Understanding how participants display in layout views](#)).

## Understanding conference views

When you join a conference, you will have a set of available layout options from which you can choose to display the conference participants. Typically you can choose from two primary groupings of layouts:

- ▶ **Same-size panes** — in these formats, all conference participants display in the same size pane on the video screen and thus have the same level of focus or importance. For example, a conference with four participants might display each in a pane sized to be a quarter (1/4) the size of your video screen.
- ▶ **Variable-size panes** — in these formats, conference participants display in various pane sizes on the video screen depending on their "importance". For example, a layout might display the participant who is speaking in a pane larger than the other conference participants. Or, you might choose to focus on a particular participant (see [Selecting participants](#)).

## Choosing a conference view

To switch among available conference views:

1. Change the camera control to "far".
2. Use the up/down navigation keys to toggle through the available format options.
3. Refer to the table below for assistance interpreting the icons that appear on-screen.

Icon	Icon description
	You are scrolling up through the layout views.
	You are scrolling down through the layout views.
	You have stopped scrolling through the layout views.

## Selecting participants

When viewing a conference with variable-size panes (see [Variable-size panes](#)), you can choose which participant to display in the larger panes on the video screen; when viewing a conference with equal-sized panes, you can choose which participant displays in the top left pane. You do this by selecting the participant following these steps:

1. Change the camera control to “far”.
2. Select a layout view.
3. Use the left/right navigation keys to scroll through the focused participants.
4. Refer to the table below for assistance interpreting the icons that appear on-screen.

Icon	Icon description
	You are scrolling through the participant list in a counter-clockwise direction.
	You are scrolling through the participant list in a clockwise direction.
	Speaking participant has the focus.

## Understanding participants' status

During the conference, various icons might appear in the pane of specific participants. Refer to the table below for assistance interpreting these icons.

Icon	Icon description
	This participant has been given priority in the layout views. A participant is made "important" using controls on the MCU web pages.
	The audio quality from this participant is poor.
	The audio quality from this participant is good.

Icon	Icon description
	The video quality from this participant is poor.
	The video quality from this participant is good.

## Using far-end camera controls

While in a conference, you might need to change the camera settings for one of the conference participants. For example, if you want to zoom in on a particular speaker in a large group, or if you cannot see the speaker. To do this:

1. Change the camera control to “far”.
2. Select the largest displayed participant pane.
3. Press Zoom. The far-end camera control icon appears, and you can now control the far-end camera.
4. Refer to the table below for assistance interpreting the icons that appear on-screen.

Icon	Icon description
	You are now controlling the viewing angle of the far-end video camera.
	You are moving the remote far-end camera down.
	You are moving the far-end video camera up.
	You are moving the far-end video camera to the left.
	You are moving the far-end video camera to the right.
	You are zooming in with the far-end video camera.
	You are zooming out with the far-end video camera.

## Understanding in-conference status icons

During the conference, various icons might appear in top left of the conference display. The appearance of these icons is controlled on the **Settings > Conferences** page (refer to [Configuring global conference settings](#)). Refer to the table below for assistance interpreting these icons.

Icon	Icon description
	There are both encrypted and unencrypted participants in this conference or there are streaming viewers of this encrypted conference.
	This conference is being recorded.
	There are audio-only participants in this conference. The number of audio-only participants is displayed next to the icon.
	There are streaming participants in this conference.

## The zoom out menu

The zoom out menu provides access to a number of in-conference options. Access the following options by pressing zoom out if you are using FECC, or 7 on a DTMF keyboard (on an endpoint that does not have an FECC channel):

- ▶ widescreen mode (whether this option is available or not depends on the **Transmitted video resolutions** setting on the **Settings > Conferences** page and on the capabilities of the endpoint)
- ▶ borders
- ▶ mute audio
- ▶ mute video
- ▶ clear text chat (this option only appears if there is text chat on the screen)
- ▶ return to auto attendant (this option only appears if the participant joined the conference from the auto attendant)

## The conference welcome message

You can configure a welcome message on the MCU and a duration for that message. Participants joining a conference on the MCU will see the message displayed at the bottom of their endpoint's display. After the configured message duration has elapsed, the message will be removed.

The **Conference welcome message** controls are on the **Settings > Conferences** page (refer to [Configuring global conference settings](#)).

The welcome message will only be displayed to streaming viewers if the duration is configured as *permanent*.

If you change the welcome message when there are active participants, any participants' currently displayed welcome messages will only change if the new message is configured as *permanent*.

## Being invited into a conference

Depending on how your video endpoint is configured, an incoming video call might come through to your regular phone or to your video endpoint (see [Calling into conferences](#) for a brief description of the difference). Note that even if a video call arrives on your regular phone, the video portion of the call will appear on your associated video endpoint (if one is configured).

As with any other type of incoming call, you can choose to answer the call or not.

### Being invited into a conference using ConferenceMe

The conference administrator can invite you to join a conference using ConferenceMe. The invite will be a http:// link included in an email.

Clicking the link loads a web browser that allows you to load ConferenceMe and be connected directly to your conference. If you do not already have ConferenceMe installed, you can access the installer from this page before joining the conference.

See [Using ConferenceMe to join conferences](#) for more information.

# Understanding how participants display in layout views

The default behavior of the Cisco TelePresence MCU is to display the "loudest" participants in the most prominent layout panes. If there are more contributors than there are panes available, then the "quietest" participants are not shown.

However, different styles of layout introduce slight subtleties to this behavior, and in addition there are a few ways in which participants or conference administrators may change the system used for pane assignment. In particular, you may want to set the [pane placement](#) for a conference yourself.

In this section:

- ▶ [Big panes vs. small panes](#)
- ▶ [Participants viewing themselves](#)
- ▶ [Changing view focus](#)
- ▶ ["Important" participants](#)
- ▶ ["Important" participants combined with view focus](#)
- ▶ [Clipped panes](#)
- ▶ [Video streams vs. fixed bitmap images](#)
- ▶ [Muted participants](#)
- ▶ [Automatic lecture mode](#)
- ▶ [H.243 floor control](#)

## Big panes vs. small panes



This layout can be seen as the most traditional video-conferencing view. Each participant is displayed in the same sized pane as the other participants. If there are more than four participants, the four most significant (the four loudest) are displayed, with one pane each.



By contrast, these example layouts have some larger panes, and the participants shown in those panes are seen as more significant than the other contributors. When allocating participants to panes, the MCU always fills the largest panes first. If there are more participants than panes then there will never be empty big panes and non-empty small panes.

To reduce the number of view changes when different participants speak (for example, when people change from being active speakers to inactive contributors), the MCU duplicates participant views for layouts with more than four small panes. For the three example layouts shown above, the participant(s) shown in the large pane(s) of the first two layouts will be duplicated in the surrounding small panes. However, the four small panes of the third layout will show different participants to those displayed in the three big panes. This reduces the impact of audio volume changes on the composed layout while not needlessly wasting view space. However, it is possible to configure the MCU not to duplicate participant views in this way if so desired (see [Configuring global conference settings](#)).

## Participants viewing themselves

When considering which participants to show in which panes, a participant's self view has the lowest priority. This has two main implications:

### ► Participant pane selection

When choosing participants to display, the MCU considers the viewer last. This prevents the participant who is the active speaker from seeing only themselves. In this case, while everyone else will see the active speaker, the active speaker will see the previous active speaker in their biggest view pane.

### ► View family layout selection

When the MCU is required to choose a layout from a view family, it does so based on the number of video contributors to the conference. However, when calculating the number of video contributors for a particular view, the MCU does not consider any video stream being received from the viewer.

Thus, with five participants in a conference and everyone seeing the standard equal-sized view family (2x2, 3x3 or 4x4), each of the five contributing participants will always see the 2x2 view with themselves excluded. However, streaming viewers will see the conference display using the 3x3 view with five panes in use.

You may configure the MCU never to show participants their own video stream in small panes (see [Configuring global conference settings](#)). If this is the case, then participants viewing layouts with some panes larger than others will never see their own video stream in any of the small panes, even if there are free slots. They may still appear in large panes, for example if the view focus is manually changed to show their video.

## Changing view focus

Using the tilt (up and down) Far End Camera Controls on a connected video endpoint causes the view to change, cycling through the available families and then the conference-wide or per-participant custom layouts (if enabled).

In addition, the pan (left and right) Far End Camera Controls on a connected video endpoint can be used to focus the view on a particular participant, as opposed to the MCU allocating participants to panes based solely on the volume of the audio being received from them.

To reduce the disruption of the view when cycling through conference participants, there is a short delay after selecting a new focused participant before the rest of the view layout reverts to the "correct" arrangement of participants in panes.

## "Important" participants

For each conference, one active participant can be set as "important". This means that the MCU considers this participant first when deciding which contributors to show in which layout panes, rather than their position in the list being set by how loudly they are speaking. See the [Control](#) setting in the [conference participant list](#).

## "Important" participants combined with view focus

Both "Changing view focus" and "Important participants" above involve a single specific participant being shown in the biggest pane available, even if that participant is not currently the loudest speaker. This causes a potential conflict, which is resolved dependant on the type of layout.



In this type of view (a layout in which all panes are of equal size), the focused participant is shown in the first pane, the one at the top left. The important participant is shown in the "next" pane, normally the one immediately to the right of the first pane.



This type of layout displays a single big pane and several small panes. If the view focus has been changed (for instance with left and right Far End Camera Control operations), then the big pane shows the selected participant. If a conference participant has been set to "important" then its video stream is shown in the big pane.

However, if a participant has been set to "important" *and* the view is focused on a (different) specific participant, the focused participant is shown in the big pane, and the "important" participant is shown in one of the small panes.

If the view has been focused on a participant and the same participant is also set as "important" then that participant is shown in the big pane.



These layouts have two large panes and several smaller ones. If the view focus has been changed (for instance with left and right Far End Camera Control operations), then the upper or left large pane shows the focused participant. If a participant has been set to "important" then that video stream appears in the lower or right large pane.

If the same participant is focused *and* "important", that video stream appears in the upper or left pane.



For these layouts, if the view has been focused on a particular participant, that participant appears in the upper or left large pane. If a participant has been selected as "important", that participant will be shown in the lower or right large pane.

In layouts with three large panes, even if the view is focused on a specific participant and another has been set to "important", one large pane remains. This pane displays the "loudest" remaining participant.

## Clipped panes

The majority of the conference layouts defined by the Cisco TelePresence MCU, for example:



have in common that all of their panes, whether big or small, have the same aspect ratio as the view itself. That is, the panes' widths and heights are in the same proportion as the width and height of the encompassing view.

By comparison, there are some defined conference layouts, for example:



in which this aspect ratio preservation does not occur. In these cases, the MCU scales the participant video stream according to the larger dimension of the pane.

For example, in the layout to the left, the size of the top left pane is one half of the view width and two thirds of the view height. Because two thirds is greater than one half, the MCU scales the participant video stream to two thirds of its size and thus a small amount of the left and right of the original image will not appear in the final composed layout.

## Video stream vs. fixed bitmap images

For video conference participants, the image that displays in the layout view pane is either the live video stream (if viewing from the video endpoint) or a captured video image from the current video stream (if viewing from the web interface).

However, audio-only participants do not have any associated video to display. Instead, you can assign a fixed image (in bitmap format) to a specific participant. When the participant joins a conference as an audio-only participant, this image appears in the layout pane. To enable this feature, the participant must be added as a user to the MCU, have an associated E.164 telephone, and have a designated image file uploaded. See [Adding and updating users](#).

## Muted participants

### Audio mute

Participants who have had their audio muted from the web interface do not contribute audio to the conference. Additionally, muted participants are not considered when the MCU calculates the loudest speakers to display in the largest panes, even if the participant had previously been in one of those positions.

Note that other participants will not have an indication that a participant has been muted. They simply will no longer hear that participant speaking.

### Video mute

Participants who have had their video muted from the web interface do not contribute video to the conference. They will continue to contribute audio as normal, unless it is muted separately.

## Automatic lecture mode

Automatic lecture mode allows a lecturer to be shown in full-screen view to the students. In this mode, the lecturer will continue to see the normal (continuous presence) view. That is, the lecturer will see the students and not himself. When **Automatic lecture mode** is enabled for a conference, the MCU identifies the loudest speaker as the lecturer. If you have configured a custom layout for the lecturer, that will override the normal continuous presence view. For the other participants (the students), the view of the lecturer (the loudest speaker) overrides any custom layout. You can enable automatic lecture mode for scheduled and/or ad hoc conferences either in the conference configuration or in the relevant conference template.

## H.243 floor and chair control

Some H.323 endpoints support a feature known as floor and chair control that is encompassed by the H.243 protocol. This is not currently supported by SIP.

The MCU supports the following H.243 features:

- ▶ a participant can "take the floor" in a conference. On "taking the floor" their video contribution is sent to all conference participants as a "1 x 1 view" (full-screen view). If the active floor (temporarily or permanently) has no video channel established to the MCU then endpoints will see their "normal" continuous presence view; if there is a video channel from the active floor participant, everyone will see that video (except for the person who currently has the floor), and this will override any view family or custom layout setting
- ▶ a participant can "take the chair". On "taking the chair", a participant can:
  - nominate a "broadcaster"; that is, they can choose which participant's video will be sent to all other participants in "1 x 1 view" (full-screen view)
  - decide to disconnect any other participant(s)

Note that the ability of a participant to "take the chair" is affected by how they joined the conference. A participant who joined the conference as a guest will not be able to "take the chair".

- ▶ an endpoint can receive the names of the other endpoints in the same conference. Different endpoints act on this in different ways.

Whether or not these features are supported in a conference depends on the individual conference settings (**Conference > Add conference**) and the global **H.243 floor and chair control setting** (**Settings > Conference**). Refer to [Adding and updating conferences](#) and [Global conference settings](#).

Where a conference supports floor and chair control, or floor control only:

- ▶ the MCU will advertise the ability to handle H.243 when establishing (and receiving) H.323 connections
- ▶ any H.243-capable endpoint can request the floor, and all endpoints (be they chairperson or guest) will be granted it as long as no other endpoint in that conference has already done so

Where the conference supports floor and chair control:

- ▶ any H.243-capable endpoint can request the chair, and any participant who has joined the conference as a chairperson will be granted it as long as no other endpoint in that conference has already done so

If an active participant in a conference has taken the chair or the floor, it is indicated in the status column of the **Participant list** page.

If you change the **Floor and chair control** for a conference currently taking place, there will be no immediate effect. That is, an existing floor or chair participant will not have that status removed.

For ad hoc conferences, you can alter the **Floor and chair control** setting, which is *Allow floor control only* by default, through that conference's configuration page when it is active.

# Displaying conference lists

The **Conference List** displays information about active, scheduled, and completed conferences. To access this list, choose **Conferences**.

By default, the **Conference List** is accessible by all users (even those who have not logged into the MCU). However, administrators can disable public access to this list. To do so, go to **Settings > User interface**.

On this page:

- ▶ [Active conferences](#)
- ▶ [Scheduled conferences](#)
- ▶ [Completed conferences](#)

## Active conferences

Active conferences are currently in progress. The following information is displayed for each conference:

Field	Field description	Usage tips
<b>Name</b>	The name of the conference, which is either the name entered when the conference was scheduled, or, in the case of certain specialized types of conferences, a name chosen automatically by the MCU when created.	Specialized conference types are described below in <b>Description</b> .  Click the conference name to display detailed information about the conference and participants.
<b>Streaming</b>	Displays the conference streaming page. This option is not necessarily available for every conference; some users do not have access to other user's conferences. Some conferences may require a PIN to access this feature.	This option is only available if you have streaming enabled on your MCU. To enable streaming, go to <b>Settings &gt; Streaming</b> .  See <a href="#">Using streaming to view conferences</a> for more information.
<b>ConferenceMe</b>	Displays the ConferenceMe page. The ConferenceMe page allows users to join the conference using the ConferenceMe client. If they do not already have ConferenceMe installed on their computer, they can access the installer from the ConferenceMe page.	This option is only available if you have ConferenceMe enabled on your MCU. To enable ConferenceMe, go to <b>Settings &gt; Streaming</b> .  See <a href="#">Using ConferenceMe to join conferences</a> for more information.
<b>Description</b>	Additional information about the conference, which can assist users joining conferences.  You can add the description when scheduling a conference. If you do not add a description or the conference has not been scheduled in advance, the description displays one of the following:  ▶ <i>&lt;scheduled&gt;</i>	

Field	Field description	Usage tips
	<p>The conference has been scheduled in advance using the MCU web interface, but the owner has not entered a description.</p> <ul style="list-style-type: none"> <li>▶ <i>&lt;ad hoc&gt;</i> The conference was created dynamically during an auto attendant session and will end when the last participant using the auto attendant exits the conference.</li> <li>▶ <i>&lt;auto attendant&gt;</i> This type of conference indicates that a participant is currently connected to the auto attendant and navigating the menus.</li> </ul>	
<b>Owner</b>	The configured owner of the conference.	See <a href="#">Conference ownership</a> for additional information.
<b>Registration</b>	<p>The status of a conference with respect to its H.323 gatekeeper and/or SIP registration. Depending on the conference settings, there is a maximum of four registrations for each conference: H.323 numeric ID, H.323 guest numeric ID, SIP numeric ID, and SIP guest numeric ID. The Registration field will show failed if any of the registrations has not completed successfully.</p> <ul style="list-style-type: none"> <li>▶ <i>n/a</i> This conference is not configured to be registered with a gatekeeper or SIP registrar; because of this, there is no applicable registration status to show</li> <li>▶ <i>Registering</i> This conference is in the process of registering with the gatekeeper or SIP registrar</li> <li>▶ <i>Failed</i> At least one of the registrations for this conference has failed</li> <li>▶ <i>Registered</i> All IDs associated with this conference have been registered successfully with the gatekeeper / SIP registrar</li> <li>▶ <i>Disabled</i> One or more of the IDs associated with this conference has been configured to be registered with the gatekeeper or SIP registrar, but that registration has not been attempted due to another setting taking precedence. For H.323 calls this</li> </ul>	<p>If the MCU can connect to an H.323 gatekeeper, each numeric ID (for both chairperson and guest privileges) for a conference can be registered with that gatekeeper as a different directory number. This allows H.323 users to dial directly into a particular conference (with the correct privileges) instead of connecting first to the MCU's auto attendant and navigating the menu system.</p> <p>Likewise, if the MCU can connect to a SIP registrar, each conference can be registered with that registrar using either (or both) Numeric ID and Guest ID. This allows SIP users to dial directly into a particular conference (with the correct privileges) instead of connecting first to the MCU's auto attendant and navigating the menu system. Note that for SIP, unlike H.323, the conferences must be configured on the SIP registrar before the MCU can register them.</p> <p>There is a maximum of four registrations for each conference: H.323 numeric ID, H.323 guest numeric ID, SIP numeric ID, and SIP guest numeric ID. To view further details about the registrations for a conference, click the conference name and then the Statistics tab to view the conference's <a href="#">Statistics</a> page.</p> <p>For tips on configuring gatekeepers, see <a href="#">Configuring H.323 gatekeeper settings</a>.</p> <p>For tips on configuring SIP registrars, see</p>

Field	Field description	Usage tips
	<p>might occur if, on <b>Settings &gt; Gatekeeper</b>, either the <b>H.323 gatekeeper usage</b> is disabled or the <b>Allow numeric ID registration for conferences</b> option is unselected. For SIP calls this might occur if, on <b>Settings &gt; SIP</b>, the <b>Allow numeric ID registration for conferences</b> is unselected</p> <p>Note that when there is a problem with the registration, the status is a link to the conference's <a href="#">Statistics</a> page.</p>	<p><a href="#">Configuring SIP settings.</a></p> <p>For more information about configuring conferences (and chairperson and guest IDs and PINs), refer to <a href="#">Adding and updating conferences.</a></p>
<b>Participants</b>	<p>The number of currently active contributors to the conference. This value does not include streaming viewers or any other endpoints that are viewing only and not contributing any media to the conference.</p> <p>If a limit on the number of participants has been set for the conference, the <b>Participants</b> value is shown as <i>A / B</i>, where <i>A</i> is the number of active participants and <i>B</i> is the configured limit. If it has not been possible to reserve all of the required ports for a conference (for instance because of a configuration error), this value will display in red as an error indication.</p>	<p>If a conference is protected by a PIN and you are not logged in as the administrator, the number of participants is hidden until the PIN is entered. In this case, the <b>Participants</b> value displays as <i>&lt;PIN required&gt;</i>.</p>
<b>Start time</b>	<p>When the conference began. If the conference started before today, the date also displays.</p>	
<b>Time remaining</b>	<p>How long the conference still has to run. If the conference does not have a limited duration, this column displays as <i>&lt;forever&gt;</i>.</p>	

## Scheduled conferences

Scheduled conferences are either in progress or are yet to start. You can review this list and make some changes to it:

- ▶ To remove a scheduled conference, select the conference via its associated check box and click **Delete selected**.
- ▶ To schedule a new conference, click **Add new conference** (see [Adding and updating conferences](#)).

The MCU supports up to 200 scheduled conferences.

Field	Field description	More information
<b>Name</b>	The name of the conference chosen when the conference was scheduled.	Click the conference name to display detailed information about the configuration of the conference and, if it is currently

Field	Field description	More information
		active, its participants.
<b>Numeric ID</b>	The number that you can dial to join the conference.	
<b>Security</b>	Whether a PIN is required to join the conference.	
<b>Owner</b>	The configured owner of the conference.	See <a href="#">Conference ownership</a> for additional information.
<b>Status</b>	Whether a conference is: <ul style="list-style-type: none"> <li>• <i>Yet to start</i> The conference's configured start time has not yet arrived.</li> <li>• <i>In progress</i> The conference is running and is available for video conferencing endpoints to join. A scheduled conference in this state will also appear in the <b>Active conferences</b> list.</li> <li>• <i>Awaiting repeat</i> The conference is not currently running, but has been previously active and is now waiting to be re-activated when the time of its next repetition is reached.</li> </ul>	There is no explicit status for "finished" – when a conference is not in progress and is not scheduled to become active again then it is moved to the <b>Completed conferences</b> list.
<b>Start time</b>	When the conference began. If the conference started before today, the date also displays.	
<b>End time</b>	When the conference will end.	

## Completed conferences

Completed conferences have finished and are not scheduled to repeat.

- ▶ To remove specific conferences from the list, select the conferences via their associated check boxes and click **Purge selected**.
- ▶ To remove all conferences from the list, click **Purge all**.

Field	Field description	More information
<b>Name</b>	The name of the conference chosen when the conference was scheduled.	Click the conference name to display detailed information about the configuration of the conference.
<b>Owner</b>	The configured owner of the conference.	See <a href="#">Conference ownership</a> for additional information.
<b>Start time</b>	When the conference began. If the conference started before today, the date also displays.	

Field	Field description	More information
<b>End time</b>	When the conference ended.	

# Adding and updating conferences

The information required to add or update a conference is nearly identical. Refer to these topics for details:

- ▶ [Adding a conference](#)
- ▶ [Updating a conference](#)
- ▶ [Adding configured endpoints](#)
- ▶ [Using IDs and PINs](#)

## Adding a conference

To add a conference:

1. Go to **Conferences > Add new conference**.
2. Complete the fields referring to the table below for the most appropriate settings for the conference. Note that the defaults that appear on the **Add conference** page are controlled by the conference's template, which you can alter (see [Using conference templates](#)).
3. Click **Add conference** to add the conference and return to the **Conference List**. The recently added conference appears either in the **Active** or **Scheduled Conferences** depending on its scheduled start time.

Ad hoc conferences (if you allow them) are added by users in the auto attendant.

## Updating a conference

To update an existing conference:

1. Go to **Conferences**.
2. Click a Conference name and then click the **Configuration** tab.
3. Edit the fields referring to the table below for the most appropriate settings for the conference.
4. Click **Update conference** to add the conference and return to the **Conference List**. The updated conference appears either in the **Active** or **Scheduled Conferences** depending on its scheduled start time.

## Ad hoc conferences

The following settings can be updated for existing ad hoc conferences:

- ▶ PIN
- ▶ Numeric ID registration
- ▶ Floor and chair control
- ▶ Automatic lecture mode
- ▶ Visibility
- ▶ Encryption
- ▶ Layout control via FECC/DTMF
- ▶ Mute in-band DTMF

- ▶ Allow DTMF \*6 to mute audio
- ▶ Mute on join
- ▶ Streaming
- ▶ Content channel video
- ▶ Content contribution from endpoints
- ▶ Transmitted content resolution

See the table below for more information:

Field	Field description	Usage tips
<b>Parameters</b>		
<b>Name</b>	The name that users will see on auto attendant screens and on the MCU's web interface.	<p>Conference names must be unique; conferences cannot share names.</p> <p>Only scheduled conferences have a configurable <b>Name</b>. Ad hoc conferences use the 'conference number' entered by the participant who has created the conference as the <b>Name</b> and <b>this is not configurable after the conference has been created.</b></p>
<b>Description</b>	Additional information about the conference, which can assist users joining conferences.	<p>Use the description to provide more detailed information about the conference than the name alone conveys.</p> <p>This is an optional field for scheduled conferences; ad hoc conferences cannot be given a description.</p>
<b>Numeric ID</b>	The unique identifier used for dialing in to the conference (as a chairperson participant) using an auto attendant or through an H.323 gatekeeper or SIP registrar. This number can also be used to stream the conference from the Streaming-only interface. For more information about chairpersons and guests, refer to <a href="#">Using IDs and PINs</a> .	<p>When connected to an auto attendant, participants can join a conference by typing its numeric identifier.</p> <p>If you plan to allow audio-only participants, then you will need to enter either a Numeric ID or a Guest numeric ID.</p> <p>If H.323 gatekeeper registration is enabled for a conference, the MCU attempts to register the conference with an E.164 telephone number, which is comprised of the <a href="#">Registration prefix</a> and the numeric identifier.</p> <p>If SIP registration is enabled for a conference, then the <b>Numeric ID</b> is registered with the SIP registrar.</p> <p>Conferences that are simultaneously active must not share a <b>Numeric ID</b>. For example, a conference on a Tuesday and a conference on a Thursday can share a <b>Numeric ID</b>, whereas two permanent conferences cannot share a <b>Numeric ID</b>. The same number can be used for the</p>

Field	Field description	Usage tips
		<p><b>Guest numeric ID</b>, if there are two different PINs. Additionally, because the numeric identifier is used in gatekeeper registration, conferences and auto attendants cannot share a numeric identifier value.</p> <p>For more information, refer to <a href="#">Using IDs and PINs</a>.</p> <p>For ad hoc conferences created via the auto attendant, the number allocated by the conference creator becomes the <b>Numeric ID</b>. If ad hoc conferences are registered with the gatekeeper and /or SIP registrar, participants can dial in using this number. Note that the actual number that H.323 participants will dial depends on whether prefixes are used in the <b>Settings &gt; Gatekeeper</b> page.</p> <p>You cannot configure the <b>Numeric ID of an ad hoc conference</b>; the <b>Numeric ID</b> of an ad hoc conference is set by the conference creator as the "conference number" at the time the conference is created.</p>
<b>PIN</b>	Provides a level of security to conference access.	<p>If a conference has a PIN set, users cannot join the conference or change its configuration without entering the correct PIN. Depending on the conference settings, it may be possible for participants to join a conference as a chairperson (using the Numeric ID and PIN), or as a guest (using the Guest numeric ID and Guest PIN).</p> <p>For an ad hoc conference, you can configure a PIN both at the time of conference creation and also while the conference is running. You can also force ad hoc conference creators to use a PIN (controlled on the <b>Settings &gt; Conferences</b> page).</p>
<b>Guest numeric ID</b>	The unique identifier used for dialing in to the conference (as a guest participant) using an auto attendant or through an H.323 gatekeeper or SIP registrar. This number can also be used to stream the conference from the Streaming-only interface. For more information about chairpersons and guests, refer to <a href="#">Using IDs and PINs</a> .	<p>When connected to an auto attendant, participants can join a conference by typing its Guest numeric ID.</p> <p>If you plan to allow audio-only participants, then you will need to enter either a Numeric ID or a Guest numeric ID.</p> <p>If H.323 gatekeeper registration is enabled for a conference, and you have entered a Guest numeric ID, the MCU attempts to register the conference with an E.164</p>

Field	Field description	Usage tips
		<p>telephone number, which is comprised of the <a href="#">Registration prefix</a> and the Guest numeric ID.</p> <p>If SIP registration is enabled for a conference, and you have entered a Guest numeric ID, then the <b>Guest numeric ID</b> is registered with the SIP registrar.</p> <p>Conferences that are simultaneously active must not share a <b>Numeric ID</b>. For example, a conference on a Tuesday and a conference on a Thursday can share a <b>Numeric ID</b>, whereas two permanent conferences cannot share a <b>Numeric ID</b>. The same number can be used for the <b>Guest numeric ID</b>, if there are two different PINs. Additionally, because the numeric identifier is used in gatekeeper registration, conferences and auto attendants cannot share a numeric identifier value.</p> <p>For more information, refer to <a href="#">Using IDs and PINs</a>.</p> <p>Ad hoc conferences cannot be configured with <b>Guest numeric IDs</b> or <b>Guest PINs</b>.</p>
<b>Guest PIN</b>	Provides secure access to conferences for guest participants.	<p>If a conference has a PIN set, users cannot join the conference or change its configuration without entering the correct PIN. Participants joining as guests have restricted privileges. For more information, refer to <a href="#">Using IDs and PINs</a>.</p> <p>Ad hoc conferences cannot be configured with <b>Guest numeric IDs</b> or <b>Guest PINs</b>.</p>
<b>Owner</b>	The owner of the conference, usually the user ID of the user account that the person who scheduled the conference logged in with.	<p>You may or may not be able to change the conference owner, depending on your privilege level. See <a href="#">Conference ownership</a> for additional information.</p> <p>This setting does not apply to ad hoc conferences.</p>
<b>Numeric ID registration</b>	<p>Enables the MCU to attempt to register the <b>Numeric ID</b> and/or <b>Guest numeric ID</b> with the configured H.323 gatekeeper and/or SIP registrar.</p> <p>To globally enable the MCU to allow conferences to register to the SIP registrar, go to <b>Settings &gt; SIP</b> and select Allow numeric ID registration for conferences.</p> <p>To globally enable the MCU to allow</p>	<p>This setting applies to both the Numeric ID and Guest numeric ID (if you have set both). For more information, refer to <a href="#">Using IDs and PINs</a>.</p> <p>For ad hoc conferences, whether or not they are registered with the gatekeeper and/or SIP registrar depends on the <b>Numeric ID registration</b> setting on the <b>ad hoc conferences template</b>. You can edit the Numeric ID registration setting for</p>

Field	Field description	Usage tips
	<p>conferences to register to the gatekeeper, go to <b>Settings &gt; H.323</b> and select Allow numeric ID registration for conferences.</p>	<p>individual active ad hoc conferences. For more information about templates, see <a href="#">Using conference templates</a>.</p> <p>For scheduled conferences, whether or not they are registered with the gatekeeper and/or SIP registrar depends on the configuration of the individual conference's <b>Numeric ID registration</b> setting.</p>
<p><b>When only guests remain</b></p>	<p>Controls what happens to the conference when the last participant with chairperson status leaves the conference. The options are:</p> <ul style="list-style-type: none"> <li>▶ <i>Disconnect all participants</i>: this is the default option. When the last participant with chairperson status leaves the conference, all other participants will be disconnected</li> <li>▶ <i>Take no action</i>: all participants may continue the conference until the last one disconnects</li> </ul>	<p>This setting applies to scheduled conferences that include guest participants (that is, those who have joined the conference using the Guest numeric ID (and PIN if one was set)).</p>
<p><b>Floor and chair control</b></p>	<p>Controls "Floor and chair control" settings for this conference. The options are:</p> <ul style="list-style-type: none"> <li>▶ <i>Do not allow floor or chair control</i>: the use of floor and chair controls is not allowed in this conference</li> <li>▶ <i>Allow floor control only</i>: only floor control is allowed in this conference; chair control is not allowed. Any participant can 'take the floor' so long as no other participant has currently 'taken the floor'</li> <li>▶ <i>Allow floor and chair control</i>: both floor and chair control are allowed in this conference. Any participant can take the floor, and any chairperson participant can take the chair so long as no other participant has currently done so</li> </ul>	<p>Some H.323 endpoints support a feature known as floor and chair control that is encompassed by the H.243 protocol. For more information, refer to <a href="#">H.243 floor and chair control</a>.</p> <p>If you change the <b>Floor and chair control</b> setting for a conference currently taking place, there will be no immediate effect. That is, an existing floor or chair participant will not have that status removed.</p> <p>If the unit-wide <b>Floor and chair control</b> setting on the <a href="#">Global conference settings</a> page is set to <i>Disabled</i>, it will not be possible to use floor or chair control operations in any conference.</p> <p>For ad hoc conferences, you can configure the <b>Floor and chair</b> setting through the ad hoc conferences template. You can also edit the <b>Floor and chair</b> setting for active ad hoc conferences.</p>
<p><b>Automatic lecture mode</b></p>	<p>Select if you want the loudest speaker to appear full screen to the other participants in a conference. If the conference requires this functionality, choose how quickly the loudest speaker will appear in full-screen view to the other participants. Choose from:</p> <ul style="list-style-type: none"> <li>▶ <i>Immediately</i>: As soon as the MCU identifies a participant as the loudest speaker, the MCU will show that</li> </ul>	<p>This setting is most useful where the conference is a lecture.</p> <p>Automatic lecture mode allows a lecturer to be shown in full-screen view to the students. In this mode, the lecturer will continue to see their normal (continuous presence) view. That is, the lecturer will see the students and not himself.</p>

Field	Field description	Usage tips
	<p>participant in full screen to the other participants in the conference. The loudest speaker will continue to see their normal continuous presence conference view. If another participant interrupts the loudest speaker, that participant becomes the loudest speaker and will be seen in full screen by the other participants (and that participant will see their normal continuous presence view).</p> <ul style="list-style-type: none"> <li>▶ <i>After 10 seconds:</i> When the MCU identifies a participant as the loudest speaker, if that participant continues to be the loudest speaker for 10 seconds then the MCU will show that participant in full screen view to the other participants. The loudest speaker will continue to see their normal continuous presence conference view. If another participant interrupts the loudest speaker, everyone will immediately see their normal continuous presence view. If the interrupter continues to speak for 10 seconds, the MCU identifies that participant as the loudest speaker who will then be shown in full screen to the other participants (and the new loudest speaker will continue to see their normal continuous presence view).</li> <li>▶ <i>After 30 seconds:</i> As for <i>After 10 seconds</i>, but the MCU waits for 30 seconds before showing the loudest speaker in full screen view.</li> <li>▶ <i>After 1 minute:</i> As for <i>After 10 seconds</i>, but the MCU waits for one minute before showing the loudest speaker in full screen view.</li> <li>▶ <i>Disabled:</i> The loudest speaker will not be shown in full screen view to the other participants. All conference participants will see the normal continuous presence conference view, or a custom layout if one has been specified.</li> </ul>	<p>The MCU identifies the lecturer as being the loudest speaker.</p> <p>If you have configured a custom layout for the lecturer, that will override the normal continuous presence view. For the other participants (the students), the view of the lecturer (the loudest speaker) overrides any custom layout.</p> <p>Note that you can use the conference's <b>Mute on join</b> setting together with the settings for the lecturer's endpoint's <b>Initial audio status</b> to ensure that the full screen view does not get needlessly interrupted.</p> <p>If you disable <b>Automatic lecture mode</b>, this change will take effect when the MCU identifies a new loudest speaker.</p> <p>For ad hoc conferences, you can configure <b>Automatic lecture mode</b> through the ad hoc conferences template. You can also edit the <b>Automatic lecture mode</b> setting for individual active ad hoc conferences.</p>
<b>Visibility</b>	<p>Indicates the visibility of the conference on the auto attendant and the web interface. The options are:</p> <ul style="list-style-type: none"> <li>▶ <i>Public:</i> the conference will be listed in the auto attendant and be visible to all users of the web interface</li> <li>▶ <i>Private:</i> the conference will not be listed in any auto attendant except for</li> </ul>	<p>For private conferences not visible on an auto attendant, participants will still be able to join the conference if they know the PIN.</p> <p>Note that only admin users can choose which conferences are visible on a given auto attendant.</p> <p>For ad hoc conferences, you can configure</p>

Field	Field description	Usage tips
	<p>auto attendants specifically set to show it. The conference will also only be visible in the web interface to the conference owner and to the admin user</p>	<p><b>Visibility</b> through the ad hoc conferences template. You can also edit the <b>Visibility</b> setting for individual active ad hoc conferences.</p> <p>For more information, refer to <a href="#">Adding and updating an auto attendant</a>.</p>
<b>Encryption</b>	<p>The encryption setting for this conference, if you have the encryption feature key enabled.</p>	<p>If encryption is enabled unit-wide (through the <b>Settings &gt; Encryption</b> page), you can set one of:</p> <ul style="list-style-type: none"> <li>▶ <i>Required:</i> encryption must be used for this conference</li> <li>▶ <i>Optional:</i> encryption is optional for this conference</li> </ul> <p>For ad hoc conferences, you can configure encryption through the ad hoc conferences template. You can also edit the <b>Encryption</b> setting for individual active ad hoc conferences.</p> <p>This setting is grayed-out if encryption is disabled on the <b>Settings &gt; Encryption</b> page.</p> <p>Note that to be able to use encryption, the Encryption feature key must be present on the MCU.</p>
<b>Layout control via FECC / DTMF</b>	<p>Prevents or permits conference participants changing their view layout or focused participant using Far-end Camera Control (FECC) or DTMF tones. Choose from:</p> <ul style="list-style-type: none"> <li>▶ <i>Disabled:</i> in this conference, participants will not be allowed to change their view layout using either FECC or DTMF, unless you have overridden this setting in an endpoint's individual configuration</li> <li>▶ <i>FECC only:</i> in this conference, participants will be only be allowed to change their view layout using FECC, unless you have overridden this setting in an endpoint's individual configuration</li> <li>▶ <i>FECC with DTMF fallback:</i> in this conference, participants will be allowed to change their view layout using FECC. If FECC is not available, this participant will be able to use DTMF</li> </ul>	<p>You may wish to prevent participants from changing their view layout in a managed conference, or classroom environment.</p> <p>This is a per-conference option, but you can still configure <b>Layout control via FECC / DTMF</b> on a per-participant basis.</p> <p>For ad hoc conferences, you can configure <b>Layout control via FECC / DTMF</b> through the ad hoc conferences template. You can also edit this setting for individual active ad hoc conferences.</p> <p>When calling out to an endpoint, if you have configured <b>Layout control via FECC / DTMF</b> for that endpoint, it will override this setting (otherwise the endpoint will use the conference configuration for this setting).</p> <p>This setting will apply to endpoints which connect to the conference via an auto attendant or by dialing in directly.</p>
<b>Mute in-band DTMF</b>	<p>Set the option for the muting of in-band DTMF sent from endpoints in this conference. Note that this sets the</p>	<p>This is a per-conference option, but you can still enable or disable the muting of in-</p>

Field	Field description	Usage tips
	<p>conference configuration for this option, but you can also override it for individual endpoints in the conference (through the configuration of the individual endpoints). Choose from the following (the setting will be applied to all endpoints in the conference that are configured to <i>use the conference configuration for Mute in-band DTMF</i>):</p> <ul style="list-style-type: none"> <li>▶ <i>Never</i>: the in-band DTMF will never be muted. Any DTMF tones sent from these endpoints will be audible to conference participants.</li> <li>▶ <i>Always</i>: the in-band DTMF will always be muted. Any DTMF tones sent from these endpoints will not be audible to conference participants.</li> <li>▶ <i>When used for MCU control</i>: if a participant is using in-band DTMF to control conference layout and for other in-conference features, the tones will be muted and will not be audible to conference participants. The MCU will only expect a participant to use DTMF for MCU control if <b>Layout control via FECC / DTMF</b> is set to <i>FECC with DTMF fallback and FECC is unavailable to the endpoint</i>.</li> </ul>	<p>band DTMF on a per-participant basis.</p> <p>In some circumstances, you might need to override this setting for individual endpoints. For example, where a conference is cascaded onto an audio bridge, it might be useful for one of the participants in that conference to be able to send in-band DTMF to the MCU. This is for the purposes of sending the conference ID or PIN to the audio conferencing bridge. In this case, the <b>Mute in-band DTMF</b> setting for the endpoint of that participant needs to be <i>Never</i> (and you configure this in an endpoint's configuration, see <a href="#">Configuring an H.323 endpoint</a> and <a href="#">Configuring a SIP endpoint</a>).</p> <p>For ad hoc conferences, you can configure the <b>Mute in-band DTMF</b> setting through the ad hoc conferences template. You can also edit the <b>Mute in-band DTMF</b> setting for individual active ad hoc conferences.</p>
<p><b>Allow DTMF *6 to mute audio</b></p>	<p>Choose between:</p> <ul style="list-style-type: none"> <li>▶ <i>Enabled</i>: When this setting is enabled, a participant in this conference can mute their audio by typing *6 into their DTMF (number) keypad.</li> <li>▶ <i>Disabled</i>: For participants in this conference, pressing *6 on an endpoint's number keypad achieves nothing.</li> </ul>	<p>This can be useful for participants calling into a conference on a traditional, audio-only phone. Sometimes there is no other control for muting audio on these phones.</p> <p>When enabled, participants can turn off their microphones (mute their audio) by pressing *6; pressing *6 again turns the microphone on again.</p> <p>For ad hoc conferences, you can configure the <b>Allow DTMF *6 mute audio</b> setting through the ad hoc conferences template. You can also edit it for individual active ad hoc conferences.</p>
<p><b>Invite pre-configured participants</b></p>	<p>Indicates when the MCU should invite any pre-configured endpoints into a conference. The options are:</p> <ul style="list-style-type: none"> <li>▶ <i>At the start of the conference</i> Pre-configured participants will be called as soon as the conference starts.</li> <li>▶ <i>When at least one other participant is present</i> Pre-configured endpoints will only be</li> </ul>	<p>Select which option fits your requirements best. Calling pre-configured endpoints <i>at the start of the conference</i> is most appropriate for repeating conferences with a particular start time.</p> <p>Calling pre-configured endpoints <i>when at least one other participant is present</i> is most appropriate for permanent conferences; such conferences are typically un-attended for much of the time,</p>

Field	Field description	Usage tips
	called after at least one other participant joins the conference.	<p>and it may only be useful to invite pre-configured endpoints when others are present.</p> <p>This setting only applies to scheduled conferences.</p>
<b>Mute on join</b>	Select to initially mute audio and/or video from all participants when they join the conference.	<p>If you select to mute audio on join, when a participant dials-in to a conference, he will not be able to contribute audio to the conference. If you select to mute video on join, when a participant dials-in to a conference, he will not be able to contribute video to the conference. You can mute both audio and video if required.</p> <p>This can be altered during the course of the conference by going to <b>Conferences</b> and clicking on the name of a conference and altering a participant's settings. For more information about altering a participant's settings during a conference, refer to <a href="#">Viewing the conference participant list</a>.</p> <p>Note that for pre-configured participants and for participants to whom the MCU dials out, the <b>Initial video status</b> and <b>Initial audio status</b> settings will be used to determine whether or not audio and video contributions are muted. For more information about dialing out to participants, refer to <a href="#">Adding participants</a>. For more information about pre-configured participants, refer to <a href="#">Configuring an H.323 endpoint</a> and <a href="#">Configuring a SIP endpoint</a>.</p> <p>For an ad hoc conference, you can configure the <b>Mute on join</b> setting through the ad hoc conferences template. You can also edit the <b>Mute on join</b> setting for active ad hoc conferences.</p>
<b>Streaming</b>	Allow unicast and multicast streaming access to individual conferences.	<p>For an ad hoc conference, you can configure <b>Streaming</b> through the ad hoc conferences template. You can also configure <b>Streaming</b> for active conferences.</p> <p>To use streaming, it must be enabled unit-wide on the <b>Settings &gt; Streaming</b> page. See <a href="#">Configuring streaming settings</a> for additional information.</p>
<b>Content channel video</b>	If <i>Enabled</i> , this conference is able to support an additional video stream, sent potentially to all connected endpoints, intended for showing content video.	For ad hoc conferences, whether or not content channel video is allowed is controlled by the <b>Content channel video</b>

Field	Field description	Usage tips
	<p>This content video is typically high resolution, low frame rate data such as a presentation formed of a set of slides. Such presentation data can be sourced by:</p> <ul style="list-style-type: none"> <li>▶ an endpoint specifically contributing a separate content video stream,</li> <li>▶ the MCU being configured to use an endpoint's main video stream as the conference's content channel,</li> <li>▶ a dedicated VNC connection,</li> <li>▶ or user-generated markup data (the markup is normally, though not always, used in conjunction with content video or VNC data).</li> </ul>	<p><b>setting in the ad hoc conference template.</b></p> <p>To use content, it must be enabled unit-wide on the <b>Settings &gt; Content</b> page. See <a href="#">Configuring content settings</a> for additional information on MCU-wide content configuration parameters.</p>
<p><b>Content contribution from endpoints</b></p>	<p>Whether, by default, endpoints are permitted to contribute the content channel for a conference through the mechanism of opening a content video channel.</p> <p>There can only be one endpoint contributing content video at any one time, and the MCU arbitrates between them. Therefore, even with this parameter set to <i>Enabled</i>, the ability of the endpoint to contribute content video will be affected by other endpoints' behavior.</p>	<p>If this setting is <i>Disabled</i>, it is still possible to enable content contribution on a per-endpoint basis when the conference is active. Similarly, it is possible to disable content contribution from specific endpoints, either while they are connected or via their configuration. For more information about endpoint configuration, refer to <a href="#">Configuring H.323 endpoints</a> and/or <a href="#">Configuring SIP endpoints</a>.</p> <p>This setting only applies to scheduled conferences. For ad hoc conferences, whether or not endpoints are allowed to contribute content is controlled by the <b>Content contribution from endpoints</b> setting in the ad hoc conference template.</p> <p>To use content, it must be enabled unit-wide on the <b>Settings &gt; Content</b> page. See <a href="#">Configuring content settings</a> for additional information on MCU-wide content configuration parameters.</p>
<p><b>Transmitted content resolutions</b></p>	<p>Choose the resolution for the content channel that will be transmitted to endpoints in this conference:</p> <ul style="list-style-type: none"> <li>▶ <i>4:3 resolutions only</i>: the MCU will encode the content and transmit it in a resolution of ratio 4:3</li> <li>▶ <i>16:9 resolutions only</i>: the MCU will encode the content and transmit it in a resolution of ratio 16:9</li> <li>▶ <i>Allow all resolutions</i>: the MCU will decide on the most optimal resolution depending on information about capabilities sent by the endpoints in the conference</li> </ul>	<p>For ad hoc conferences, transmitted content resolution is controlled by the ad hoc conference template.</p>

Field	Field description	Usage tips
<b>Maximum video participants</b>	<p>When the MCU is not in port reservation mode, this parameter sets a limit on the number of endpoints which can connect to the conference as video participants.</p> <p>A participant counts (as a single unit) towards the video limit whether the MCU is sending a video stream to that participant or a video stream is being received.</p>	<p>If you do not want to limit the number of participants who can join this conference and use video, leave this field blank.</p> <p>This field is only shown if the MCU is not in port reservation mode.</p> <p>This field only applies to scheduled conferences.</p>
<b>Maximum audio-only participants</b>	<p>When the MCU is not in port reservation mode, this parameter sets a limit on the number of endpoints which can connect to the conference as audio-only participants.</p> <p>A participant counts (as a single unit) towards the audio limit whether the MCU is sending an audio stream to that participant or an audio stream is being received.</p>	<p>If you do not want to limit the number of participants who can join this conference to use just audio, leave this field blank.</p> <p>This field is only shown if the MCU is not in port reservation mode.</p> <p>This field only applies to scheduled conferences.</p>
<b>Video ports to reserve</b>	<p>In port reservation mode, this parameter specifies the number of video ports to reserve.</p> <p>A participant counts (as a single unit) towards the video reservation value whether the MCU is sending a video stream to that participant or a video stream is being received.</p>	<p>This value is both a reservation and a limit; the MCU guarantees that this many video participants can connect to the conference, but no more than this will be able to join.</p> <p>This field is only shown if the MCU is in port reservation mode.</p>
<b>Audio-only ports to reserve</b>	<p>In port reservation mode, this parameter specifies the number of audio-only ports to reserve.</p> <p>A participant counts (as a single unit) towards the audio reservation value whether the MCU is sending an audio stream to that participant or an audio stream is being received.</p>	<p>This value is both a reservation and a limit; the MCU guarantees that this many audio-only participants can connect to the conference, but no more that this will be able to join.</p> <p>This field is only shown if the MCU is in port reservation mode.</p>
<b>Start time</b>		
<b>Start time</b>	The time at which the conference will begin.	
<b>Start date</b>	The date on which the conference will begin.	
<b>Set to current time</b>	Sets the conference start time to the current time on the MCU.	The current time on the MCU is determined by the settings in the <b>Settings &gt; Time</b> page, which can only be modified by the administrator. See <a href="#">Displaying general status</a> for additional information.
<b>Duration</b>		
<b>Permanent</b>	Allows you to retain a conference and its settings for an infinite period of time.	
<b>Maximum duration</b>	Limits the duration of the conference for one instance of the conference.	These fields are not available or necessary for conferences set to <i>permanent</i> .

Field	Field description	Usage tips
<b>Repetition</b>		
<b>Interval</b>	<p>Which days and / or weeks the conference repetitions will occur.</p> <p>The repetitions will always start at the same time of day: the conference's configured <b>Start time</b> (see above), and will last for the same amount of time: the configured <b>Maximum duration</b>.</p>	<p>The start date is taken into account when determining when the first repetition should occur. For instance, if the start date is a Wednesday and the conference is scheduled to repeat every Monday, Tuesday and Wednesday then it will occur only on Wednesday in the first week and on all three specified days in subsequent weeks.</p>
<b>Termination</b>	<p>If a conference is set to repeat, its repetitions can be configured to go on forever, stop after a certain date, or to occur only a certain number of times.</p>	<p>The first activation of a conference counts as a "repetition", so configuring a conference to repeat but terminate after 1 repetition is equivalent to it not repeating at all.</p>

## Adding configured endpoints

You can choose to pre-configure endpoints to be part of a conference. These endpoints will be automatically invited into the conference by the MCU. This is useful if you regularly invite the same participants into a conference. To select which previously configured endpoints will be pre-configured for this conference, press **Pre-configured participants**. (This button may also show a number in parentheses to indicate the number of participants that are currently pre-configured). Refer to [Adding participants](#) for more details.

## Using IDs and PINs

There are two types of conference participant: chairperson and guest. IDs and PINs allow participants to connect to conferences as the correct participant type.

Chairperson participants use **Numeric ID** and optionally, **PIN**; guest participants use **Guest numeric ID** and optionally, **Guest PIN**.

A conference will not begin until the first chairperson joins. This means that guests will see a black screen/hear silence with on screen text 'Waiting for conference chairperson' and an audio prompt after five seconds and then every minute thereafter. Streaming viewers will see a black screen/hear silence although mark-up and text chat will work.

You can control the behavior when the last chairperson leaves the conference (that is the **When only guests remain** setting). The two options are:

- ▶ all participants are disconnected (default)
- ▶ all participants may continue the conference until the last one disconnects (*take no action*)

If you set both **Numeric ID** and **PIN**, you must set either:

- ▶ both **Guest numeric ID** and **Guest PIN**

or

- ▶ neither **Guest numeric ID** nor **Guest PIN**

If you use a **Guest PIN**, you must use a **PIN**. If you use a **Guest numeric ID**, you must use a **Numeric ID**.

Note that for a conference **Numeric ID** and **Guest numeric ID** can be the same only if the two PINs are different; likewise, both PINs can be the same if the IDs are different.

Note that participants dragged and dropped into a conference on the web interface will be chairperson participants.

Where no IDs or PINs are configured for a conference, all participants will be chairpersons regardless of how they join.

### **H.323 and SIP registration**

Both Numeric ID and Guest numeric ID can be registered with an H.323 gatekeeper and/or SIP registrar to enable participants to dial in to conferences directly and as the correct participant type. The **Numeric ID registration** setting applies to both IDs.

### **Audio-only participants**

Audio-only participants can be guest or chairperson participants by connecting to a conference using either a Guest or Chairperson ID. In the case of an audio-only guest, if no chairperson has yet joined the conference, they will hear an audio prompt informing them of that. The conference will start when the first chairperson joins.

## Using conference templates

Templates control the default settings for conferences. Currently there are two templates:

- ▶ The **top level** template: provides the settings that appear on the **Add conference** page which you see when you add a scheduled conference by using the web interface. When you add a conference, you can then alter any of the settings. You can alter some conference settings during the conference.
- ▶ The **ad hoc conferences** template: provides the configuration for all ad hoc conferences. You can alter some conference settings during the conference.

If a template has a parent, it can use the settings from the parent or override them with its own values.

For information about the conference configuration settings, refer to [Adding and updating conferences](#) which lists all conference configuration settings.

# Conference ownership

Each scheduled conference (i.e. conferences that are configured via the web interface with a start time and, optionally, a duration and repetition) has an associated *owner*. This owner is the ID of a configured user, and normally corresponds to the user who scheduled the conference.

## Scope of conference ownership

Conference ownership affects only web interface control of conferences - in particular, it plays no part in validating video conferencing endpoints' attempts to join conferences when they connect to the MCU via H.323 or SIP. Restricting conference entry in this situation is accomplished via conference (or auto attendant) PINs, as before.

## User privileges

The actual implications of conference ownership depend on the privilege of the user; specifically:

Privilege level	Effects of conference ownership
<ul style="list-style-type: none"> <li>• <b>administrator</b></li> <li>• <b>conference creation and full control</b></li> </ul>	Users with these privilege levels are able to create and own conferences, and are able to exercise full control of all conferences.
<ul style="list-style-type: none"> <li>• <b>conference creation and limited control</b></li> </ul>	Users with these privilege levels are able to create and own conferences. They have full control of conferences they own, and limited control of conferences owned by other users.
<ul style="list-style-type: none"> <li>• <b>conference creation</b></li> </ul>	Users with these privilege levels are able to create and own conferences. They have full control of conferences they own, but no control of conferences owned by other users.
<ul style="list-style-type: none"> <li>• <b>conference detail</b></li> <li>• <b>conference list plus streaming</b></li> <li>• <b>conference list only</b></li> </ul>	Users with these privilege levels are not able to own conferences or change any conference's configuration.

## Levels of conference control

As described above, a user privilege level confers a certain level of control over a conference, with that level of control possibly depending on whether that user is the conference owner or not. These conference control levels have the following meaning:

Conference control level	Description
<ul style="list-style-type: none"> <li><b>full control</b></li> </ul>	<p>This level of control permits the following operations:</p> <p><b>Participant control</b> With full conference control, a user is able to disconnect participants, connect new participants to the conference, and end the conference whenever they want to.</p> <p><b>Configuration access</b> A user with full conference control can view and modify any aspect of the conference's configuration. This includes the start time, end time, or repetition characteristics, whether streaming is enabled, and which endpoints are pre-configured as participants.</p> <p><b>Changing live conference parameters</b> When the conference is in progress, a user with <i>full control</i> is permitted to send a text message to all connected participants' video displays and change the <a href="#">Conference custom layout</a>.</p> <p>Additionally, full control includes all of the operations covered by <i>limited control</i>, detailed below.</p>
<ul style="list-style-type: none"> <li><b>limited control</b></li> </ul>	<p>This level of control permits the following operations:</p> <p><b>Viewing the participant list</b> The participant list shows the names of the endpoints currently connected to the conference, a summary of that endpoint's status and, if they are a video participant, a thumbnail preview image of the video stream they are supplying. Thumbnail previews are shown by default, but you can configure the user interface not to show them. The setting that controls this is the <b>Show video thumbnail images</b> option on the <b>Settings &gt; User interface</b> page. Note that the MCU will not show thumbnail previews on the participant list page if encryption is required for a conference. However, thumbnail previews will be shown for conferences where encryption is optional and there are encrypted participants.</p> <p>Via the participant list, it is also possible to mute (or stop muting) individual endpoints' audio, change the conference's <a href="#">"important participant"</a>, and enable or disable the participants' ability to affect their own layouts via Far-End Camera Control or DTMF tones.</p> <p><b>Controlling video sent to participants</b> This includes being able to choose what view layout (or family) is used for constructing the conference view being sent to a participant, changing the <a href="#">Participant custom layout</a>, and whether to send widescreen or standard format video to that endpoint.</p> <p><b>Controlling participant cameras</b></p>

Conference control level	Description
	<p>The web interface can be used to send control commands (e.g. pan and tilt) to a remote participant's camera.</p> <p><b>Viewing audio status</b> Waveforms of audio channels being sent to, and received from, the participant can be viewed, audio gain applied, and participant audio can be muted if required.</p> <p><b>Sending messages to individual participants</b> Textual messages can be sent to an individual participant, and will appear on their displays.</p> <p><b>Viewing participant statistics and diagnostics</b> This allows details of the media streams being sent to, and received from, participants to be viewed, plus endpoint-specific characteristics to be examined.</p>
<ul style="list-style-type: none"> <li>• <b>no control</b></li> </ul>	<p>This means that none of the above operations are possible. Depending on the specific privilege level, either the list of scheduled conferences will not be shown to the user, or the list will be shown but conferences over which the user has no control will be presented as names rather than hyperlinks.</p>

## Changing the owner of a conference

A user whose privilege level is either *conference creation and full control* or *administrator* is also able to change the owner of a conference. A conference owner can be changed to either a user with conference creation rights or to "none", signifying that no user should be considered the owner of that conference. Also, when scheduling new conferences, users with these privileges are able to choose which owner is initially associated with the conference.

## Deleting users who are conference owners

If a user account is deleted, any conferences owned by that user have their owner reset to "none".

# Reservation of MCU media ports

The MCU is able, if required, to allocate its available media ports in advance to specific conferences. This means that it is able to guarantee that a certain number of participants will be able to join that conference, irrespective of how many other people are using the MCU for other conferences at the same time.

## Media port types

There are three types of media ports available on the MCU, **video ports**, **audio-only ports**, and **streaming and content channel ports**.

For information about the number and type of ports provided by each MCU model, refer to [MCU port matrix](#).

The term *video port* refers to a port that can be used by a video-conferencing endpoint for a call. Thus, a video port includes both video **and** audio streams (bidirectionally) and so the number of video ports available represents the number of "normal" video calls that the MCU is able to maintain simultaneously.

## Participant port usage

In general, each endpoint in a conference is able to use either a video port or an audio-only port, though normally the MCU will assign video ports to video-capable devices and audio-only ports to audio-only devices.

If a video-capable device joins a conference which has just audio-only ports available, the MCU will assign it an audio-only port – that participant will be able to listen to other people and contribute their own audio to the conference but the MCU will not transmit video to it (and will not use any video received from it). If an audio-only device such as a simple telephone joins a conference which has just video ports available, the MCU will assign it a video port, which includes audio capability. The video capability of that allocation will not be used, but the audio device will be able to participate as normal in the conference. The exception to this is VNC - because this is a video-only protocol, the MCU does not permit VNC connections to use audio-only ports.

## Streaming

Streaming a conference requires use of a **video port** or a **streaming and content channel port**. Where streaming and content channel ports are provided, streaming viewers and conferences' content channel (H.239 or BFCP) video allocations will use the streaming and content channel ports rather than the available video ports; where streaming and content channel ports are not provided, streaming viewers and content channel allocations will use available video ports.

If a streaming and content channel port or a video port is unavailable (or not allocated in advance when the MCU is in *Reserved mode*), it will not be possible to stream that conference. If a video port has been allocated for streaming a conference, any number of streaming viewers will be able to view that conference via streaming, at any combination of available bit rates.

## MCU media capacity

The total number of media ports available depends on the MCU model; refer to the product datasheets available on the web site, or to [MCU port matrix](#) for more information.

## Configuring the MCU

How MCU media ports are allocated, and which options and settings are available, is controlled by the **Media port reservation** setting on the [Settings > Conferences](#) page.

### Unreserved mode

This is the mode that the unit runs in when the **Media port reservation** setting is configured as *Disabled*, and is the mode that the MCU uses by default. With this scheme, you can specify a maximum value for the number of video and audio-only ports each conference is allowed to use on the Conference configuration page. These limits are optional, and by default there is no configured limit.

The configured limits are strictly *maximum* values; in particular, setting such a limit does not guarantee that that many participants will be able to join the conference. It is perfectly possible to set these values such that the sum of the configured limits across all active conferences exceeds the total number of ports available on the MCU.

### Reserved mode

This is the mode that the unit runs in when the *Media port reservation* setting is configured as *Enabled*. With this scheme, each conference must be configured with a number of video ports to reserve and a number of audio-only ports to reserve. These values differ from the maximum port values set in [Unreserved mode](#) in a number of ways:

- ▶ **Reservations are guaranteed**  
As well as being maximum values (i.e. enforcing limits on the number of conference participants), port reservation values also guarantee that that many endpoints are able to participate in the conference.
- ▶ **Port reservations are mandatory**  
In unreserved mode, it is not necessary to specify a number of video or audio-only ports for a conference. In reserved mode, however, every conference must have configured reservations for both video and audio-only ports.
- ▶ **Over-allocation is not permitted**  
Port reservations guarantee that a certain number of participants will be able to join a conference; because of this, the MCU will not permit these reservations to be configured such that the total number of reserved ports at any given time exceeds the total number of ports available. See [Clashing reservations](#) for additional information.

### Clashing reservations

In order to honor configured port reservations, the MCU must ensure that at any given time the number of reserved ports does not exceed the total media capacity. This entails some level of *clash detection* when you schedule conferences change their configuration.

The MCU considers two conferences to be clashing if they can ever be active simultaneously. When validating a conference schedule, the MCU looks at the maximum number of ports reserved by other conferences which can be active at the same time, and checks that the number of ports requested by the conference being added or changed is guaranteed to be available. If, for instance, the MCU has 20 video ports available in total, it will not be possible to set up two conferences which require 15 video ports each if they are scheduled such that they ever overlap.

In the simple case of conferences which start at specific times and end at specific times (or, indeed, are permanent), it is easy to see whether they clash. The more complex cases involve repetition, and it is important to bear in mind that port reservations are only permitted when the MCU can guarantee

them for **every** repetition of a conference. As an example, a conference scheduled to run from 08:00 to 10:00 on the second Monday of each month will be deemed to clash with a conference configured to run from 09:00 to 09:30 every Monday, even though the former will only really clash with the latter every fourth or fifth week.

In general, to make best use of the available MCU media ports, you should not schedule conferences to be longer than needed, and you should limit repetitions, either by end date or number, wherever possible.

## Ad hoc conferencing

Because port reservations are mandatory in **Reserved mode** every active conference must have configured values for the number of video ports and the number of audio-only ports to reserve for it. In turn this means that every active conference must be configured, and therefore *ad hoc conferences* are not permitted when in **Reserved mode**.

This affects the operation of the MCU in the following ways:

- ▶ **Auto attendant usage**  
In reserved mode, the [Create new conference](#) option will not be shown on video auto attendant menus, even for auto attendants configured to display this option.
- ▶ **Auto attendant configuration**  
When configuring new or existing [auto attendants](#) via the web interface, the *Creation of new conferences* parameter will not be available.
- ▶ **Calls to unknown E.164 number configuration**  
This setting on the [Conference settings](#) page also offers a *Create new conference* option. This is not available in reserved mode and becomes equivalent to the *Disconnect caller* option.

## Auto attendant connections

If a participant calls in to the MCU and connects to an auto attendant, the MCU does not know which conference they will join until they make a selection from the auto attendant menu.

In [Unreserved mode](#), the auto attendant connection allocates a media port from those not currently in use. If all of the media ports are in use, the endpoint's connection will be dropped by the MCU.

In [Reserved mode](#), the auto attendant connection effectively "borrows" a media port from those not currently in use. However, this borrowed media port has a lower priority than a media port used by a conference participant, and if the auto attendant connection "borrows" the last remaining media port, then that connection will be dropped if another endpoint connects directly to a conference and requires a reserved media port.

## Changing MCU port reservation mode

In general, changing port reservation mode when there are active connections is not recommended. The effects of changing mode include, but are not necessarily limited to:

- ▶ **Destruction of ad hoc conferences**  
Any ad hoc conferences in progress will be destroyed when changing to port reservation mode and their participants dropped because [ad hoc conferences](#) are not permitted in port reservation mode.

► **Participant disconnection**

Participants in a scheduled conference may be disconnected. For each conference, the maximum port usage values (for unreserved mode) and the ports to reserve (for reserved mode) are configured and stored separately. Therefore when changing port reservation mode, it is possible that there are more active participants than allowed in the particular mode. In this case, participants are disconnected to reduce the number to that allowed.

# Content channel video support

The Cisco Telepresence MCU supports an additional video channel known as the *content channel* for each conference. This feature encompasses:

- ▶ H.239 video streams sent from the MCU to viewing H.323 endpoints
- ▶ Sourcing the content channel from a H.323 endpoint's H.239 video stream, a SIP endpoint supporting content using BFCP, or a VNC connection
- ▶ Showing the content as part of the *main video* channel, where an endpoint cannot, for whatever reason, receive the content channel as an additional video channel. That is, the participant will see the content as a pane in the conference layout
- ▶ Streaming the content channel to users' desktop machines as a pane in the conference view
- ▶ Streaming the content channel to users' desktop machines as a separate window (where markup and text chat can also be used)<sup>(\*)</sup>
- ▶ Allowing graphical and textual "markup" of the content channel <sup>(\*)</sup>
- ▶ Text chat between conference participants <sup>(\*)</sup>

\* - these features require the [web conferencing feature key](#).

## Content channel vs. main video

The H.239 protocol allows the MCU to support an additional video stream to or from each connected endpoint. Therefore, there are potentially three media streams between each endpoint and the MCU: audio, main video and content video.

SIP endpoints will always receive the content as part of the *main video* channel. That is, SIP endpoints will not receive the content channel as a separate video channel, but will see the content as a pane in the *main video* channel.

The *main video* is the normal multi-pane conference view showing participants' video streams composed within the selected layout. The differences between the content channel video and the *main video* are:

- ▶ **Single layout**  
Each participant in a conference can normally select their own individual main video layout (e.g. a 2 x 2 grid of other participants, one large focused pane plus eight smaller panes) and they are free to change this layout as many times as desired while they are connected to the conference. By comparison the content channel video always shows just a single video stream, "full screen", and each viewing endpoint will see the same stream. The stream which constitutes the content channel can change any number of times during the conference, but there can be at most one contributing stream at any given moment.
- ▶ **One channel per conference**  
Each participant's main video stream is encoded independently; this means that each endpoint can be receiving its main video stream at a different bit rate, codec, or resolution to that being sent to other participants.  
However, there is a single content video stream per-conference: the MCU sends the same bit rate and resolution to all endpoints receiving content. The bit rate and resolution used is chosen to maximize the number of viewers - for instance the resolution might be reduced if a new endpoint joins the conference and its content receive capabilities are more limited than those of the other participants. Note that in conferences that use encryption, the MCU can send encrypted and unencrypted content to different participants in the same conference (albeit with the same encryption key for every participant receiving encrypted content).

► **Differing characteristics**

The range of bit rates, resolutions and frame rates available to the MCU for sending the content channel via H.239 to H.323 video conferencing endpoints is potentially as wide as that for the main video channel. However, in general, the main video channel is used for motion video (i.e. high frame rate streams) and the content channel for less dynamic video such as an accompanying presentation - typically high resolution, low frame rate.

However, the MCU allows flexibility in terms of nominating which of the available streams forms the content channel, as well as allowing control over which endpoints are permitted to start contributing content video.

► **Uni-directionality**

For the main video channel, a video conferencing endpoint would normally be both contributing (sending) a video stream to the MCU and receiving a video stream from it.

However the content channel works differently in that an endpoint can either be sending content video or receiving content video, but not both. A given endpoint may switch between being the contributor and a viewer during the conference, but it will never be both simultaneously.

## H.323/SIP endpoints' content channel support

For H.323 endpoints, depending on the specific endpoint and how it is configured, the content video stream may be displayed on a separate screen, or the endpoint may show the main video and the content video streams side by side on the same screen.

Irrespective of its content receive capability, an endpoint may or may not be able to contribute the content channel - typically, for this to be possible it will either need a second camera or some other video input such as a VCR or "video in" connection.

Some H.323 endpoints may have no support for the H.239 protocol and the MCU does not send the content channel video to SIP endpoints. However, it is still possible for such endpoints to display the content channel - the MCU is able to show the content channel within a normal view pane in the same way as it displays other conference participants. This ability is controlled by the [unit-wide Display content in normal video channel](#) setting (see [Configuring content settings](#)).

## Content channel sources

As described [above](#), a conference's content channel as sent to the set of receiving endpoints has a single source. There are several possible content channel sources:

► **H.239 video channel**

This is the most conventional content channel behavior - a H.323 conference participant opens a H.239 channel to the MCU and contributes a video stream, such as that supplied by a second camera or an attached PC.

Because there can be at most one content channel source, the H.323 endpoint needs to make a request to the MCU, and have that request accepted, before actual content channel contribution can start. If the conference already has an active content channel (for example, another endpoint is contributing H.239 video), the new request will be rejected by the MCU - it will be necessary to wait for the active contributor to cease sending H.239 video before the new endpoint is able to start. However, if you have enabled **Automatic content handover** (on the [Settings > Content](#) page), the new request will be granted automatically.

► **BFCP video channel**

BFCP (Binary Floor Control Protocol) is a protocol that allows for an additional video channel (known as the content channel) alongside the main video channel in a video-conferencing call that uses SIP. A SIP conference participant opens a BFCP channel to the MCU and contributes a video stream, such as that supplied by a second camera or an attached PC.

Because there can be at most one content channel source, the SIP endpoint needs to make a request to the MCU, and have that request accepted, before actual content channel contribution can start. If the conference already has an active content channel (for example, another endpoint

is contributing content video), the new request will be rejected by the MCU - it will be necessary to wait for the active contributor to cease sending content video before the new endpoint is able to start. However, if you have enabled **Automatic content handover** (on the **Settings > Content** page), the new request will be granted automatically.

Note that the MCU does not send the content channel to SIP endpoints as a separate video channel, but can show the content channel within a normal view pane in the same way as it displays other conference participants. This ability is controlled by the [unit-wide Display content in normal video channel](#) setting (see [Configuring content settings](#)).

► **VNC connection**

A VNC connection is one where the MCU has made a connection to a remote device (normally a desktop PC) and is receiving a video stream from that device; this is typically used for including a slide-based presentation in a conference.

If **Automatic content handling** is disabled, and a conference is configured with content channel support, when a VNC "participant" is added to that conference, the MCU will attempt to use the VNC video as the content channel. This is normally the desired behavior; however, there are a couple of caveats:

- If there is a video conferencing endpoint actively contributing content video then it will not be possible to immediately switch over to using the VNC video instead - the switch over will occur when the participant closes his content channel.
- With more than one active VNC connection in a conference, the first will be used as the content channel source, and subsequent connections will be shown in main video layout panes, just as if the conference had no content channel facility. If the currently active VNC connection is either disconnected or its use as the content channel disabled, another VNC video stream will start to be used as the content channel.

If **Automatic content handling** is enabled:

- If there is a video conferencing endpoint actively contributing content video, the MCU will automatically switch to using the VNC video as the content channel.
- If there is already an active VNC connection in a conference, a second VNC connection will automatically replace the first as the content channel (rather than appearing as a pane in the main video).

► **Participant main video**

It is also possible for the MCU to use a endpoint's main video channel as the conference's content channel. This has the same caveats as the use of a [VNC connection](#) for the content channel, as described above.

► **Video markup**

Whichever of the content channel sources mentioned above is active, the MCU provides the facility for users to add markup to the video channel. This markup consists of an additional video *overlay* on to which graphics may be drawn and text added; content channel viewers see the content channel source video plus this overlay. The overlay can be cleared later leaving the source video unaltered.

## MCU content channel configuration

### Unit-wide configuration

At the unit-wide level, the MCU can be configured to disallow the use of conference content channels completely. If the content channel facility is enabled, the MCU can be separately configured not to allow textual or graphical markup of the content, and whether to make text chat visible to connected endpoints.

You can choose to enable encryption on the MCU. When encryption is used, the content channel will be encrypted.

For more information on these configuration parameters, see [Configuring content settings](#) and [Configuring encryption settings](#).

## Per-conference configuration

Assuming that content is enabled on the MCU unit-wide, each scheduled conference can be independently configured to allow content channel operations or not. If enabled, this has an impact on the conference's [port usage](#); if disabled, then all attempts by participants in that conference to open a content channel to the MCU will be unsuccessful.

If the MCU is configured to allow encryption, each individual conference can be configured to either require encryption or to optionally use encryption. The MCU can send either encrypted or unencrypted content to different participants in a conference depending on the capabilities of those participants' endpoints.

For more information on the conference configuration parameters relevant to the content channel, see [Adding and updating conferences](#).

## Per-participant parameters

### *Content contribution*

**Content contribution** refers to the ability of video conferencing devices to contribute the content channel video for a conference via the mechanism of opening a separate video channel, distinct from its [main video](#) stream. Specifically, this section does not deal with the use of content by the MCU when sending content channels to viewing devices or the use of other protocols such as VNC to supply the content channel video for a conference.

For a conference configured with content channel video enabled, each endpoint in that conference is either permitted or prohibited from being able to contribute content video. H.239 is the protocol used by H.323 video conferencing endpoints to supply or receive content channel video; BFCP is the protocol used by SIP video conferencing endpoints to supply content channel video. Other content channel source configurations, such as the use of a [VNC connection](#), do not depend on any H.239 or BFCP contribution parameters.

Remember that what is termed *Content contribution* is more precisely described as the ability to **start** contributing content channel video via H.239/BFCP. The nature of the H.239 and BFCP protocols used between the MCU and endpoints is such that once an endpoint has successfully become the content source for a conference, the MCU is not then able to force that endpoint to stop contributing the content channel video.

While an endpoint is supplying the content channel for a conference, it is considered to be holding the virtual *content token* for the conference. This token must be relinquished before either another endpoint can start contributing video via H.239 or BFCP or a content channel source such as VNC become active. This token is normally released by a specific endpoint operation (e.g. a "stop content" option), or by that endpoint leaving the conference. However, when **Automatic content handover** is enabled, the MCU will ask the endpoint (or computer) to return the token if there is another participant who wants to start contributing content.

When **Automatic content handover** is enabled, it allows another participant to start sending content without having to wait for the current content provider to stop sending content from his computer. In this case, the MCU will start sending the 'new' content to the participants in the conference and will ask the endpoint (or computer) that was originally providing content to return the token. **Automatic content handover** is a unit-wide configuration option on the **Settings > Content** page.

By default, participants' ability to contribute content video (technically, as above, to **start** contributing H.239 or BFCP video) is determined by the per-conference **Content contribution from endpoints** setting (**Conferences > Add new conference**).

The per-conference default **Content contribution from endpoints** setting can be overridden by individual endpoints' configuration. If such an endpoint's *Content contribution* setting is *<use conference default>* then the endpoint's ability to contribute content channel video will be determined initially from the conference setting. If the endpoint setting is *<enabled>* or *<disabled>* then this will override the conference setting and that endpoint will either always be prevented from using content, or always permitted (assuming the conference of which it is part is configured with content channel support). As well as being part of each endpoint's configuration, the **Content contribution** setting can also be specified when calling out to an endpoint by address.

Irrespective of per-conference or per-endpoint configuration parameters, if a conference is configured to allow content channel operations then it is possible to explicitly enable or disable individual conference participants' ability to use content via the web browser interface (assuming a user login with [full conference control](#)).

To change the content contribution setting for an active conference participant via the web interface, first navigate to that participant's **Display** page (go to **Conferences** and click the conference you want and then click on the name of the participant whose settings you want to change). If the conference has content enabled and the endpoint in question has content capabilities, you should be able to use one of the following controls:

- ▶  allow participant to contribute content video
- ▶  do not allow participant to contribute content video

If an endpoint's ability to contribute content video has been explicitly enabled or disabled via this mechanism, that enablement or disablement will take precedence over any current or future conference or participant configuration, even if the endpoint later moves to a different conference.

### *Use of main video as content channel*

In addition to supporting the H.239 and BFCP protocols by which endpoints in a conference can supply the content channel video, the MCU also allows a participant's [main video](#) channel to be used for the content stream. This is essentially what happens by default for [VNC connections](#) in a conference configured to allow content channel video.

As detailed above, it is not possible to force an endpoint that has [started to contribute content video](#) to relinquish the virtual token that it holds. Thus, if you select an endpoint's main video channel to be the content channel source, this will only take effect if no other endpoint is supplying the content channel video stream (whether by H.239, BFCP, or through use of its main video stream). However, if you have enabled **Automatic content handover** on the **Settings > Content** page and you select an endpoint's main video channel to be the content source, this will take effect even if there is currently another content source in the conference.

To control the use of a participant's main video as the conference content channel source, the following controls are displayed on the per-conference participant list (next to the preview image of the video stream to which they relate):

- ▶  use this participant's main video stream as the content channel (this is set by default for VNC connections)

- ▶  stop using this participant's main video stream as the content channel (revert to more conventional content channel behavior such as H.239)

In rare circumstances, if more than one participant's main video channel is configured to provide the content channel (for example where a participant configured in this way is moved to another conference where there is an existing participant providing his main video channel as the content channel), then all but the active (normally, first) one will be marked with the status:

*Content: unable to use main video as source*

You might also see this warning if there is more than one VNC connection in a conference, because, when establishing a new VNC connection, the MCU will automatically configure its main (in fact, sole) video channel to be used as the content channel source if possible. To choose between multiple potential main video channels as the content source for the conference, use the control to stop using this participant's main video channel as the conference's content channel source:



on all but the source you want to use.

## Streaming

As well as the content stream (used for sending to H.323 endpoints), the MCU also generates a proprietary format version of the content channel video which can be viewed in conjunction with PC-based video streaming. This ensures that, if desired, all participants and viewers for a conference are able to access all of its associated media.

Content channel streaming also allows participants using H.323 video conferencing endpoints without H.239 capability, or SIP endpoints to view a high resolution version of the content channel. Content channel streaming also provides some features not available via the H.239 protocol:

## Markup

"Markup" is the overlaying of graphics and text onto the content channel video; this could be used, for instance, to draw attention to a specific element of a presentation slide. Markup can only be performed through the content channel streaming interface, and is accomplished via the simple mechanism of clicking and dragging with the mouse, with extra controls for changing the drawing color or clearing the markup when its usefulness has passed.

Content channel markup also has the following characteristics:

- ▶ All content streaming viewers have equal markup capabilities. This means that either all viewers are able to perform markup on the content channel video (or clear it), or all are unable to do so - it is not possible to restrict markup to just a subset of active viewers.
- ▶ The video markup, although it can only be edited by streaming viewers, will be seen by content channel streaming viewers, by participants connected via H.323 endpoints that are receiving the content channel, and by any participants that are viewing the content as part of their main video.
- ▶ Video markup is normally used as an overlay to content channel video as supplied by a H.239, BFCP, or VNC connection. However, for a conference with content channel operations enabled, it is also possible to form a content channel comprising just the markup applied to an otherwise blank video stream. This can be accomplished simply by starting to stream the conference's content channel and performing the usual markup operations on the (empty) content channel displayed. Adding markup to such a blank channel "activates" it, and will cause a video channel to be opened to those H.323 conference endpoints which have H.239 capabilities. SIP endpoints will see the video markup as a pane in the main video.

The ability of content channel streaming viewers to perform markup is governed by the [unit-wide Markup of content channel video](#) setting on the **Settings > Content** page.

## Text chat

In parallel with, though in many senses independently of, content channel streaming, the MCU also provides a mechanism for those streaming a conference's content channel to communicate with other conference participants via text messages. Beneath the window showing the content channel video, streaming viewers are able to type messages that will be sent to all other streaming viewers, as well as see messages that other users type.

In order that users contributing text messages can be identified, each content streaming viewer has an associated user ID, and this ID is pre-pended to each message when it is sent out to other viewers' displays. If the content channel streaming has been initiated via the streaming-only interface, each user is required to supply a **Sign-in name** before streaming starts, and this sign-in is used as their text chat identifier. If streaming has been initiated via the **Stream** control on the MCU conference list, the user's web interface login ID will be used as their text chat identifier.

The text chat facility provided via web browser-based content streaming is two-way in that any content channel streaming viewer is able to both contribute text and see all messages typed by other viewers. Although there is no mechanism by which endpoints are able to contribute text chat messages, the MCU is able to display the most recent text messages within endpoints' main video channels. This is intended to be of use when a presenter is connected to an MCU conference via a video endpoint and wishes to field questions raised by (content channel) streaming viewers. In this situation, the text typed by content channel streaming viewers is overlaid on the normal, multi-pane, conference layout, though restricted to approximately the lowest 1/3 of the screen.

For streaming viewers to have the option of contributing and/or reading text messages, the **View content channel** option must be selected. This option is one of the **advanced streaming options** available to users when they select to stream a conference.

The display of text chat in endpoints' views is governed by the [unit-wide Content channel text chat](#) option which is an **Overlaid text** setting on the **Settings > Conference** page. The text chat facility itself, and display of typed text to all content channel streaming viewers' windows, cannot be disabled.

## Web conferencing feature key

Some of the above content channel features require the MCU to have been configured with the *Web conferencing* feature key. The following features are only available with the Web conferencing key:

- ▶ Streaming the content channel to users' desktop machines as a separate window (rather than as a pane in the conference view)
- ▶ Markup of content channel video
- ▶ Text chat

## Port usage

### Port reservation mode

If the MCU is operating in [reserved mode](#), enabling content for a conference requires the use of an additional video port. A single video port is needed for all content channel operations, irrespective of how many viewers there are; for example, a conference involving five video endpoints (one of which is contributing a content stream and the other four viewing it) will require six video ports - **Video ports to reserve** should be set to 5, and **Content channel video** set to "Enabled" in this specific example.

In reserved mode, a conference with content enabled will require a video port for content operations even if no current participants are actively making use of content.

Note that for the MCU 4203, there are separate ports for content and for streaming viewers. For more information, refer to [MCU port matrix](#).

## Unreserved mode

If the MCU is operating in [unreserved mode](#), enabling content for a conference works in a similar way to streaming in that it will require a port to be allocated when content channel operations are first attempted for that conference. For instance, this could be when a participant opens a content channel or a user starts viewing the content channel via their web browser. When the port is no longer needed for the conference's content channel (e.g. when the last remaining participant disconnects) the port will be released for use by future participants or conferences.

## Streaming

The streaming of the content channel is performed using the port allocated for content rather than the port allocated for streaming. This means that it is possible to stream the content channel (for example, to use the video markup feature) for conferences which do not have streaming enabled. Enabling both streaming and content for a conference will mean that two additional ports will be required for that conference, over and above the video and audio-only ports used by endpoints participating in that conference.

See [port reservation](#) for more information.

## Ad hoc conferences

All of the above-mentioned features, for instance content video streams between the MCU and video conferencing endpoints, markup of content channel video and text chat, are available for use with both scheduled and ad hoc conferences.

However, whereas for scheduled conferences the availability of content is determined by a per-conference configuration setting, for ad hoc conferences it is only possible to enable or disable content on a device-wide basis. This is accomplished via the **Content for ad hoc conferences** option on the [Content settings](#) web page — if this is "Enabled" then any ad hoc conference on the MCU may use content; if "Disabled" then none may do so.

In software versions prior to 1.4(1), it was not possible to use content channel features (including the H.239 protocol between the MCU and H.323 video conferencing endpoints) with ad hoc conferences. It remains the case that ad hoc conferences are not permitted when operating in [reserved mode](#).

## Controlling in-conference features

You can control many conference features from the MCU web browser:

- ▶ [Adding participants](#)
- ▶ [Viewing the participants list](#)
- ▶ [Customizing layout views](#)
- ▶ [Displaying conference statistics](#)
- ▶ [Sending messages to all participants](#)

## Adding participants

You can add a participant to a conference in two ways: either as a "one-off" participant in to an active conference now or by adding their endpoint to the conference configuration, in which case the endpoint will be called every time the conference occurs. Their endpoint must be pre-configured for you to be able to do this, see [Adding pre-configured participants](#).

To call an endpoint immediately you can either put in their endpoint details or select a pre-configured endpoint and call out. Their endpoint can be an H.323, SIP or VNC endpoint. Refer to the first three links below for details:

- ▶ [Adding an H.323 participant](#)
- ▶ [Adding a SIP participant](#)
- ▶ [Adding a VNC participant](#)
- ▶ [Inviting ConferenceMe participants](#)
- ▶ [Adding pre-configured participants](#)

Note that participants called by the MCU will have chairperson privileges. For more information about chairperson and guest privileges, refer to [Adding a conference](#).

Note if you have the Web conferencing feature key on your MCU, you can also add ConferenceMe participants to a conference through the **Participants list** page. For more information on how to do this, see [Viewing the conference participant list](#).

### Adding an H.323 participant

To call a participant with an H.323 endpoint in to an active conference:

1. Go to **Conferences** to display the Conference List.
2. Click a Conference name and then click **Add participant**.
  - To add a new endpoint (one that you have not added as a configured endpoint):
    - i. Select the *H.323* radio button and do one of the following:
      - For **Address**, enter the IP address, E.164 number, or H.323 alias of the participant's endpoint.
      - For **Address**, enter the number configured on an H.323 gateway for the participant and select the *H.323 gateway* from the drop down list. For more information about configuring H.323 gateways, refer to [Adding and updating gateways](#)
      - For **Address**, enter an H.323 gateway and E.164 number in the form `<gateway address>!/<E.164 number>`. This allows to you make a call via a H.323 gateway that has not been pre-configured on the MCU.
    - ii. For information about other conference settings, refer to [Configuring H.323 endpoints](#)
    - iii. Click **Call endpoint**.
  - To invite an existing (configured) H.323 endpoint, scroll down to the **Endpoints** section, select the check box next to the endpoint name and click **Add selected**.

### Adding a SIP participant

To call a participant with a SIP endpoint in to an active conference:

1. Go to **Conferences** to display the Conference List.

2. Click a Conference name and then click **Add participant**.
  - To add a new endpoint (one that you have not added as a configured endpoint):
    - i. Select the *SIP* radio button and do one of the following:
      - For **Address**, enter the IP address, or SIP URI of the participant's endpoint (in the format 1234@example.com)
      - For **Address**, enter the number registered with the SIP registrar and select **Use SIP registrar** (ensuring that you have configured a SIP registrar on the **Settings > SIP** page)
    - ii. For information about other conference settings, refer to [Configuring SIP endpoints](#)
    - iii. Click **Call endpoint**.
  - To invite an existing (configured) SIP participant, scroll down to the **Endpoints** section, select the check box next to the SIP endpoint name and click **Add selected**.

## Adding a VNC participant

To call a VNC endpoint in to an active conference:

1. Go to **Conferences** to display the Conference List.
2. Click a Conference name and then click **Add VNC**.
  - To invite a new VNC endpoint:
    - i. Enter the host name or IP address of the VNC host and the VNC server password for the computer you are adding.
    - ii. Enable **Automatic disconnection** if you want the participant to be disconnected from the conference if another participant disconnects and only participants configured to be automatically disconnected remain in the conference.
    - iii. Click **Invite**.
  - To invite an existing (configured) VNC endpoint, scroll down to the **Endpoints** section, click next to the VNC endpoint name and click **Add selected**.

## Inviting ConferenceMe participants

To invite ConferenceMe participants to an active conference:

1. Go to **Conferences** to display the Conference List.
2. Click a Conference name and then click **Invite** in the ConferenceMe section.
  - To invite ConferenceMe participants:
    - i. Choose the **Local interface** you would like the ConferenceMe client to connect to.
    - ii. Choose the **Local address** you would like the ConferenceMe client to connect to.
    - iii. Choose the **TCP service** you would like the ConferenceMe client to connect with.
    - iv. Either:
      - Copy the generated **Link** and send this to ConferenceMe users.
      - OR: Click **Invite** to send an email containing the generated **Link**.

## Adding pre-configured participants

You can choose participants whose endpoints have been configured previously to be part of a scheduled conference. These participants will be automatically invited into the conference by the MCU every time the conference runs. This is useful if you regularly invite the same participants into a conference.

To add previously configured endpoints to a conference:

1. Go to **Conferences** to display the Conference List.
2. Click a Conference name and then click the **Configuration** tab.
3. Click the **Pre-configured participants** button. The pre-configured participants page will be displayed. This page lists all the endpoints that have been configured on the MCU (see [Configuring an H.323 endpoint](#) and [Configuring a SIP endpoint](#)).
4. Select which endpoints you would like to add as pre-configured participants in this conference.
5. Press **Return to conference configuration**.
6. Make any other changes you require to the conference configuration, then click **Update conference**. (If you do not click **Update conference**, you will lose the selections of configured endpoints made in the previous steps.)

## Customizing layout views

You can select custom layouts to use for all conference participants. To use this option:

1. Go to **Conferences**.
2. Click a Conference name and then click the **Custom layout** tab.
3. Click on the layout you want to use from those shown in the **Available layouts**. There are different numbers of panes and pane configurations to choose from.  
The chosen layout is displayed enlarged to the left in the **Conference custom layout** section of the page and the *Enabled* radio button is selected.
4. If you want to select the participants who appear in a pane manually rather than letting the MCU make the most appropriate selection, click **Pane placement**.  
See [Using pane placement](#) below.
5. To have all participants see this layout when they join the conference click **Make new participants see this view**.
6. If you also want all participants to see this view now, click **Switch all participants to this view**.

The following table explains the details that display.

Field	Field description	Usage tips
<b>Conference custom layout</b>		
<b>Current status</b>	Whether a custom layout can be used for this participant.	When you click a new layout from the <b>Available layouts</b> , <i>Enabled</i> is automatically chosen with the most recent layout selected.
<b>Make new participants see this view</b>	New participants joining this conference will view the conference with this custom layout rather than one of the default views (see <a href="#">Customizing a participant's layout display</a> ).  To force all participants to use this layout, click <b>Switch all participants to this view</b> .	If a participant has chosen to use a custom conference layout (see <a href="#">Customizing a participant's layout display</a> ), their view will automatically update and switch to the new view.
<b>Available layouts</b>		
	Displays all the conference layouts that are available. Click a layout to select it and make it available.	You can only select one custom layout at a time.

### Using pane placement

To have more control over which participant appears in which pane, you can use the Pane placement function. Pane placement works on a per conference basis.

Pane placement works on the selected custom layout in the **Conference custom layout** page - see [Customizing layout views](#). The chosen custom layout is shown with the panes numbered. The largest (and therefore most important) panes have the lowest numbers. Because the largest number of panes in any custom layout is 20, there are twenty drop down lists, one per pane. If the number of panes in the chosen layout is less than this, a gray line separates the panes that are used from those that do not apply to this layout.

The first time you open the pane placement page for a conference all the panes are set to <default>. The MCU decides which participant will appear in panes with this setting. See [Understanding how participants display in layout views](#) for more information.

For each pane you can select an alternative setting:

- ▶ <blank>: no participant appears in this pane
- ▶ <rolling>: this pane shows conference participants, automatically changing at the configured "Pane rolling interval" frequency from the **Settings > Conferences** page. All participants contributing video are shown in rolling panes except for any participants explicitly placed in other panes
- ▶ <loudest speaker>: the participant who is speaking the loudest at any time appears in this pane
- ▶ <content>: this pane is reserved for the content channel (see [Content channel video support](#)). If the content channel is not used in this conference, this pane will be blank.
- ▶ <name>: the name of each pre-configured endpoint in this conference, and (if the conference is active) the names of all active participants whose endpoints were not pre-configured is displayed in the lists. If you select a specific participant (endpoint), that participant appears in this pane at all times.

To use pane placement:

1. Click *Enabled*.
2. For each pane that you want to control, select an entry from the drop down list.
3. For panes that you no longer want to control individually, select <default>.
4. Click **Update pane placement**.
5. To return to the **Custom layout** page, click **Custom layout**.

## More about pane placement

When you use pane placement, bear in mind that:

- ▶ Pane placement only applies to conference custom layouts. If they are disabled so is pane placement
- ▶ You can mix panes set to <default> with panes that have other settings
- ▶ Any panes that you configure keep their setting even if you change the custom layout view. Therefore if you configure all the panes in the 20 pane layout and then move to one with only five panes, panes 1 to 5 will have the same settings as before. Panes 6 to 20 will also keep their settings, it is just that they are not used. Therefore if you subsequently move to a layout with say 10 panes, all ten panes have their settings pre-configured. This does mean that if you have participants who need to be seen at all times, you should configure them in the lower numbered panes
- ▶ You must set up pane placement for each conference that you want to use it with
- ▶ You can set up pane placement before a conference starts and configure pre-configured endpoints in to particular panes. However you can only configure active participants whose endpoints are not pre-configured when the conference is running
- ▶ Pane placement persists over conference repetitions. If you set pane placement once, the same placements are ready to be used when the conference next repeats
- ▶ If you select a particular participant for a pane and they are disconnected for any reason, that pane appears blank
- ▶ After you set up pane placement, you can still change the layout for individual participants - and they can change their layout using the far end camera controls unless you disable this for each participant individually. See [Customizing a participant's layout display](#).
- ▶ The left and right controls on a participant's far end camera control, used to select a focused participant, have no effect when pane placement is in use
- ▶ Potentially, when pane placement is in use, a participant may appear in two panes at the same time. This happens for example if a pane is configured to show a particular participant and

another pane is configured to show the loudest speaker; each time that participant is the loudest speaker, he will appear in both panes. There is a setting on the MCU's [Settings > Conferences](#) page to control this behavior called **Loudest speaker pane placement behavior**

# Displaying conference statistics

You can display statistics about a conference and use the information to quickly see how many participants are currently in the conference.

To access this option:

1. Go to **Conferences**.
2. Click a Conference name and then click the **Statistics** tab.

If the conference is active, statistics for that conference will be displayed. If the conference is completed only "No longer active" will be displayed.

Refer to the table below for information on interpreting this information.

Field	Field description
<b>Start time</b>	When the conference started.
<b>End time</b>	When the conference will complete. This will be the time at which the maximum duration of the conference will have elapsed. This setting displays as <i>&lt;permanent&gt;</i> if it has been configured to last forever.
<b>Running time</b>	The duration of this conference.
<b>(Chairperson) gatekeeper ID</b>	The status of a conference with respect to its H.323 gatekeeper and/or SIP registrations. The possible states are:
<b>Guest gatekeeper ID</b>	<ul style="list-style-type: none"> <li>▶ <i>n/a</i> This conference is not configured to be registered with a gatekeeper or SIP registrar; because of this, there is no applicable registration status to show.</li> </ul>
<b>(Chairperson) SIP registrar ID</b>	<ul style="list-style-type: none"> <li>▶ <i>&lt;number&gt; registered</i> The conference has been registered successfully with the gatekeeper or SIP registrar and can be contacted using the number indicated.</li> </ul>
<b>Guest SIP registrar ID</b>	<ul style="list-style-type: none"> <li>▶ <i>Registering</i> This conference is in the process of registering with the gatekeeper or SIP registrar.</li> <li>▶ <i>Deregistering</i> The conference is in the process of unregistering with the gatekeeper or SIP registrar. This might occur if:                             <ul style="list-style-type: none"> <li>• Gatekeeper registration has been turned off (either for that conference only or for the entire MCU)</li> <li>• The configured gatekeeper has just been changed and the MCU is in the process of unregistering from the previous gatekeeper before registering its conferences with the new one.</li> </ul> </li> <li>▶ <i>Re-registration pending / Retry timeout</i> If the MCU fails to register a conference with the gatekeeper or SIP registrar, it enters these states temporarily before re-attempting the registration.</li> <li>▶ <i>Registration disabled</i> This ID has been specifically configured to be registered with the gatekeeper or SIP registrar, but some other configuration has overridden this, causing</li> </ul>

Field	Field description
	<p>the registration to not be attempted. This state might occur if either of the <b>H.323 gatekeeper usage</b> or <b>ID registration for scheduled conferences</b> options on the <b>Settings &gt; Gatekeeper</b> page is set to <i>Disabled</i>.</p> <ul style="list-style-type: none"> <li>▶ <i>&lt;no ID set&gt;</i> The conference is configured to register with a gatekeeper or SIP registrar, but has not had a numeric identifier set.</li> </ul>
<b>Number of audio/video participants</b>	The current number of contributing audio/video participants.
<b>Highest number of audio/video participants</b>	The largest number of contributing audio/video participants who have been in the conference at the same time.
<b>Number of audio-only participants</b>	The current number of contributing audio-only participants.
<b>Highest number of audio-only participants</b>	The largest number of contributing audio-only participants who have been in the conference at the same time.
<b>Number of streaming viewers</b>	The current number of streaming participants watching the conference.
<b>Highest number of streaming viewers</b>	The largest number of streaming participants who have been watching the conference at the same time.

## Sending messages to all participants

You can send messages to all participants in a conference simultaneously. To access this option:

1. Go to **Conferences**.
2. Click a Conference name and then click the **Send message** tab.

This message appears overlaid on each participant's view, including participants viewing the conference via streaming.

Field	Field description	Usage tips
<b>Message text</b>	Enter the message to send to all conference participants.	<p>Messages must be fewer than 256 characters, but depending on the viewing screen, messages at the higher-end of this limit might not display properly. Therefore, consider limiting messages to approximately 180 characters. Messages longer than 256 characters will not be truncated; they will not display at all.</p> <p>You can disable this setting from <b>Settings &gt; Conferences</b> (see <a href="#">Conference settings</a>).</p>
<b>Position</b>	The vertical position of the message on the conference display.	Select from the top, middle, or bottom of the conference display.
<b>Duration</b>	How long the message appears on participants' video screens.	The default setting is 30 seconds. To remove all messages before they time out, click <b>Clear message</b> .

## Managing participants

You can view detailed information about conference participants:

- ▶ [Viewing the conference participant list](#)
- ▶ [Customizing a participant's layout display](#)
- ▶ [Controlling the near-end camera](#)
- ▶ [Managing a participant's audio signals](#)
- ▶ [Creating a custom participant view](#)
- ▶ [Displaying statistics for a participant](#)
- ▶ [Sending messages to one participant](#)
- ▶ [Displaying diagnostics for a participant](#)
- ▶ [Moving a participant](#)

# Viewing the conference participant list

The **Participant list** displays information about active and previous participants in the conference. To access this list, go to **Conferences** and click a Conference name. This page explains the information available in the participant list.

In this section:

- ▶ [Conference information](#)
- ▶ [Lock conference](#)
- ▶ [ConferenceMe link](#)
- ▶ [Active participants](#)
- ▶ [All-participant controls](#)
- ▶ [Previous participants](#)
- ▶ [Pre-configured participants](#)
- ▶ [Summary information](#)
- ▶ [Watching the conference](#)

## Conference information

Above the list of participants, certain information about the conference is displayed:

- ▶ **This conference is being recorded:** If the conference is being recorded on an IP VCR, this message will appear.
- ▶ **Port usage:** Each conference may have either imposed limits on the maximum number of media ports it is able to use, or a certain number of media ports reserved for its use. This section shows the video port and audio-only port reservation or limit.
- ▶ See the full description of [Port reservation](#) for additional information.
- ▶ **Streaming:** Whether streaming is in use or not for this conference.
- ▶ **Content channel:** Whether the content channel is in use or not for this conference.
- ▶ **Encryption:** If you have the encryption feature key enabled, the encryption status of the conference will be shown, which will either be *not required* or *required*.

## Lock conference

A locked conference is one where new participants cannot dial in and you cannot start to stream the conference. Existing participants and streaming viewers will maintain their connection to the conference. If a conference is locked, an icon will indicate this on the auto attendant. The auto attendant will also display the text: 'This conference is locked'.

Note that participants that are pre-configured via the API will be able to dial in to the conference even if it is locked. This allows conferences to be restricted to specific participants (known as whitelisting).

Admin users and the conference owner can lock and unlock a conference.

When the final participant leaves, by default, the conference will be automatically unlocked by the MCU. However, you can configure the MCU to keep a conference locked even when the final participant leaves. To configure this feature, go to **Settings > Conferences > Advanced settings**.

When a conference is locked, admin users and the conference owner will still be able to connect new endpoints and disconnect participants.

To lock a conference:

1. Go to **Conferences** and click on the name of the conference you want to lock. The Conference's Participant List displays.
2. Click the **Lock conference** button in the top right-hand corner of the page.

## ConferenceMe link

If you have the Web Conferencing feature key, participants will be able to join the conference using ConferenceMe (a software endpoint). To join the conference using ConferenceMe, click **Join**.

To invite participants to join the conference using ConferenceMe, click **Invite**. The MCU automatically generates an email. Alternatively, you can copy the link in the *ConferenceMe link* text box and paste it in to an email.

For more information on ConferenceMe, refer to [Using ConferenceMe to join conferences](#).

## Active participants

Refer to the table below for details about the active participant list, which you can modify in the following ways:

- ▶ You can end the conference, forcing all participants (including unicast streaming viewers) to be dropped, by clicking **End conference**.
- ▶ For conferences that are "scheduled" rather than "ad hoc", ending the conference in this way causes any configured repetition to be cancelled. The ended conference would move from the "Scheduled conferences" list to the "Completed conferences" list. The configured duration of a completed conference reflects the actual duration of the conference rather than its original configured duration. For example, if a conference was scheduled to run from 09:00 until 10:00 (one hour) and the conference ended at 09:25, the configured duration would be changed to 25 minutes.
- ▶ You can add a new H.323 or SIP video conferencing endpoint to the conference (either creating a new endpoint or by choosing an existing one) by clicking **Add participant**. See [Adding participants](#).
- ▶ You can add a new VNC endpoint to the conference (either by creating a new endpoint or by choosing an existing one) by clicking **Add VNC**. See [Adding participants](#).

Field	Field description	Usage tips
<b>Type</b>	Indicates whether the participant's endpoint is an H.323, SIP, or VNC endpoint.	
<b>Participant</b>	Displays the name of the endpoint.	
	The following may also be displayed:	

Field	Field description	Usage tips
	<p><i>User: &lt;user id&gt;</i> If the participant has been resolved to a particular configured user (for instance by matching the participant's E.164 phone number against a configured value), then the appropriate user id will be shown here.</p>	<p>User names only display if the participant list is being viewed by an administrator, because only administrators have access to the configured user database.</p>
	<p><i>Configuration: &lt;name&gt;</i> This indicates that this participant corresponds to a pre-defined endpoint. The &lt;name&gt; shown is the endpoint's configured name.</p>	<p>Configured endpoint names only display if the participant list is being viewed by a user whose privilege level is <i>administrator</i> or <i>conference creation and full control</i>; only users with these privilege levels have access to the configured endpoint database.</p>
<p><b>Controls</b></p>	<p> Makes participant important</p> <p> Makes participant unimportant</p>	<p>When you make a participant "important", it sets this participant as the focused participant. For example, this participant is considered the loudest participant even if they are not speaking.</p> <p>Only one participant can be identified as "important", and no participant is set to "important" by default.</p> <p>This option affects the layout views for the conference and individual participants. See <a href="#">Selecting a custom participant view</a> and <a href="#">Customizing layout views</a>.</p> <p>If the content channel is made "important", one participant can still be important. On endpoints that support content, the content channel will be displayed in the content channel window and the important participant will be given the focus in the main video window. On endpoints that do not support content, the important participant will be ignored as the content channel will be given the focus.</p>

Field	Field description	Usage tips
	 Prevents other participants from hearing this participant (mute).  Allows other participants to hear this participant's audio (stop muting)  Prevents others from seeing this participant's video contribution (mute).  Allows others to see this participant's video contribution (stop muting).	<p>The presence of these controls is controlled by the <b>Participant list controls</b> setting on the <b>Settings &gt; User interface</b> page (refer to <a href="#">Customizing the user interface</a>).</p>
	 Disconnects a participant from the conference	<p>You can configure the MCU to require confirmation when someone attempts to disconnect an individual participant from a conference. This setting is on the <b>Settings &gt; User interface</b> page.</p>
<b>Status</b>	<p>Displays the time at which a participant connected to the conference. If a participant is not yet fully connected (for example, if the MCU has called a participant but it has not yet answered), then that is indicated here. The resolution of the participant's video stream is also displayed.</p> <p>The following additional indications may also be displayed:</p> <p><i>Recording</i> Indicates that the conference is being recorded.</p> <p><i>Encrypted</i> Indicates that all media streams in both directions (to and from the endpoint and the MCU) are encrypted.</p> <p><i>Rx encrypted</i> Indicates that all media streams received from this participant are encrypted.</p> <p><i>Tx encrypted</i> Indicates that all media streams sent to this participant are encrypted.</p> <p><i>Pre-configured</i> This indicates that this participant is in the conference because of a pre-configured endpoint.</p>	<p>If this column is selected as the sorting field, the listing is ordered according to when the participants connected to the conference.</p> <p>Next to the encryption indication is the AES check code. This can be used in combination with information displayed by some endpoints to check that the encryption is secure.</p> <p>See <a href="#">Adding pre-configured participants</a> for additional information on adding endpoints to conferences in this way.</p>

Field	Field description	Usage tips
	<p><i>No audio capabilities</i>  <i>No video capabilities</i>            These messages indicate that the MCU has not opened a media channel to a participant's endpoint because it has no capability to receive that type of channel. For example, if the endpoint is a simple telephone, you might expect to see "No video capabilities" shown here.</p>	
	<p><i>No common audio codecs</i>  <i>No common video codecs</i>            These messages indicate that the remote endpoint had declared media capabilities, but the MCU was not permitted to open a channel that the endpoint was prepared to receive.</p>	<p>This is most likely to occur if you have disabled one or more codecs in the "Audio codecs from MCU" or "Video codecs from MCU" configuration on the <a href="#">Advanced conference settings</a> web page.</p>
	<p><i>No common audio formats</i>  <i>No common video formats</i>            These messages indicate that the remote endpoint had declared media capabilities including codecs that the MCU was permitted to send, but that the MCU was unable to transmit the specific formats declared.</p>	<p>This could occur if, for instance, the far end's advertised receive video sizes were all 4CIF (704 x 576 pixels) or above, and the MCU was set to not transmit above CIF (352 x 288 pixels).</p>
	<p><i>No common symmetric audio codecs</i>  <i>No common symmetric video codecs</i>            Indicates that the MCU was unable to open a media channel to the endpoint because the only possible channels it would be able to open were invalid due to a symmetry clash. This clash occurs when the remote endpoint declared that it is only able to receive certain types of media if it is also sending the same format, and the format in question is one that the MCU is able to send but not receive.</p>	
	<p><i>Audio port limit exceeded</i>  <i>Video port limit exceeded</i>            These messages occur if a channel was unable to be opened solely because doing so would have exceeded the port limit. This limit may be a per-conference restriction or, for those conferences which impose no such limit, it could be that all of the MCU's available ports were in use.</p>	
	<p><i>Endpoint audio and video channels rejected</i>  <i>Endpoint audio channel rejected</i>  <i>Endpoint video channel rejected</i>            Indicates that the MCU is unable to receive the media format that the endpoint is trying to transmit.</p>	

Field	Field description	Usage tips
	<p><i>Tx:</i> <i>Rx:</i> These status rows show details of the audio and (main) video streams being sent to or received from the endpoint. <i>Tx</i> refers to media streams sent by the MCU, and <i>Rx</i> means media streams being received by the MCU.</p>	<p>If the MCU has received information that the endpoint has been muted at the far end, this will be indicated in the Rx row as "muted remotely".</p>
	<p><i>Content tx: &lt;status&gt;</i> This row (if present) shows one of two things: the characteristics of the content video channel being sent by the MCU to a participant, or the reason why no such stream is currently being transmitted. The status values that can be shown here are:</p> <p><i>Content tx: pending</i> The MCU is able to send content video to the participant but is not currently doing so. This is normally because there is no active content channel video source for the conference. This status will also be shown for a participant which <i>is</i> the content channel video source and is also capable of receiving the currently active content video stream.</p> <p><i>Content tx: disabled</i> Content is enabled for this conference, but this participant is not allowed to receive it. To enable content for this participant, configure the <i>Content video receive</i> setting for this participant's endpoint (refer to <a href="#">Configuring H.323 endpoints</a>), or use the <i>enable</i> control for an active call.</p> <p><i>Content tx: no common codecs</i> There is a unit-wide setting that determines which video codec is used by the MCU for outgoing content channels. The <i>Outgoing content video codec</i> is on the <a href="#">Settings &gt; Content</a> page.</p> <p>This message means that the endpoint is able to receive content video, but not using the same codec as the MCU is configured to transmit.</p> <p><i>Content tx: resolution mismatch</i> <i>Content tx: bit rate mismatch</i> For each conference, the MCU uses a single video stream for all outgoing content connections to endpoints. This entails considering all endpoints' receive capabilities and deciding which resolution and bit rate to send in order to maximize the number of content channel viewers. These messages mean that the MCU has</p>	<p>No <i>Content tx</i> status will be shown if the conference does not have content channel video enabled, or if the endpoint has no content capabilities.</p> <p>You may change whether the MCU is prepared to send the conference content channel stream to an endpoint using content using the <i>enable</i> or <i>disable</i> control here. If sending of content video to an endpoint is currently not allowed, you will see the <i>enable</i> option, otherwise you will see the <i>disable</i> option.</p> <p>For more information about using content, refer to <a href="#">Content channel video support</a>.</p>

Field	Field description	Usage tips
	<p>been unable to include this endpoint in the set of content viewers because of its limitations with regard to video resolution or bit rate. Note that it is possible to configure a lower limit on the bandwidth of the shared content video encoding which will cause a bit rate mismatch where an endpoint is only able to receive a low bit rate stream (refer to <a href="#">Configuring content settings</a>).</p> <p><i>Content tx: mode mismatch</i> The MCU supports only <i>presentation mode</i> for its content channels; specifically, H.239 <i>live mode</i> is not supported. This status message indicates that the endpoint is content-capable but does not support presentation mode.</p> <p><i>Content tx: encryption not possible</i> The MCU is unable to send encrypted content video to this participant.</p> <p><i>FURs ignored</i> The MCU has detected that the content link to this endpoint is experiencing problems, and has stopped sending video keyframes in response to Fast Update Requests (FURs) to avoid degrading the content video sent to other conference participants. The MCU will only enter this state if the <b>Video fast update request filtering</b> setting is <i>Enabled</i> (see <a href="#">Configuring content settings</a>).</p>	
	<p><i>Content video source (main video)</i> <i>Content video source</i> The associated participant is currently supplying the conference's content channel video.</p>	
	<p><i>Content: unable to use main video as source</i> Use of this participant's main video source as the content (content channel) source has failed. This is normally because there is already a source for the content channel; either another participant's main video channel which has been configured in the same way (VNC participants are set up this way by default) or a content video channel contributed by a connected endpoint.</p>	
	<p><i>Packet loss detected</i> This message appears if packet loss is detected between the MCU and the endpoint.</p>	<p>Click <b>View</b> to open the <b>Participant statistics</b> page and display details about the connection, for example the packet errors. See <a href="#">Displaying statistics for a participant</a>.</p>

Field	Field description	Usage tips
	<p><i>Conference chair</i> This endpoint has requested and been granted the chair.</p>	<p>For more information, refer to <a href="#">H.243 floor and chair control</a>.</p>
	<p><i>Active floor</i></p> <p>The endpoint has requested and been granted the floor and its media is being broadcast to all endpoints in the conference.</p>	<p>For more information, refer to <a href="#">H.243 floor and chair control</a>.</p>
	<p><i>Inactive floor</i></p> <p>This endpoint believes it has been granted the floor, but its media is not being shown to all endpoints in the conference.</p>	<p>This situation can occur when an endpoint requested the floor, was granted the floor, and was then moved to another conference where there was already an active floor participant.</p> <p>For more information, refer to <a href="#">H.243 floor and chair control</a>.</p>
	<p><i>Assigned floor</i></p> <p>This endpoint's media is being broadcast to all endpoints in the conference, although it did not request the floor itself.</p>	<p>This situation can occur if another endpoint, while chair, has issued H.243 commands to make that endpoint the floor.</p> <p>For more information, refer to <a href="#">H.243 floor and chair control</a>.</p>
	<p><i>Cascade link to master</i> <i>Cascade link from slave</i> Indicates that this connection is a cascade link to or from another conference.</p>	
<b>Preview</b>	<p>Displays a sample still video capture of the participant.</p>	<p>Click the picture to update it.</p> <p>Note that conferences that are configured to require encryption do not show previews. Also note that previews can be disabled on the <b>Settings &gt; User interface</b> page (refer to <a href="#">Customizing the user interface</a>).</p>
	<p><i>Controls</i></p> <p>These controls are only available if the conference is configured with content enabled, and if the participant is contributing a main video channel.</p> <p> This control causes the MCU to attempt to use the participant's main video channel as the conference's content channel source. This will not be possible if the conference already has an active content channel source (either an endpoint's content video channel or another participant's main video activated via this control).</p>	<p>See <a href="#">Content channel video support</a> for additional information on H.239 and BFCP.</p>

Field	Field description	Usage tips
	 <p>This control causes the MCU to stop attempting to use the participant's main video channel as the conference's content channel source. It is necessary to use this control to switch to using a different endpoint's primary video channel or to enable content video contribution from endpoints.</p>	

## All-participant controls

Although you may use the controls described above to manage one participant at a time, for example to mute that participant's audio, or to make them important, you may wish to manage all participants at once. The **all-participant** controls permit you to do this. Note that you may continue to use the per-participant controls in conjunction with the all-participant controls. Refer to the table below for details of the controls available:

Field	Field description	Usage tips
<b>Importance</b>	 <p>Makes all participants unimportant</p>	If no participants are currently important, then this control will be unavailable.
<b>Mute</b>	 <p>This stops all audio in the conference (mute).</p>  <p>Allows all participants to hear the audio in the conference (stop muting).</p>  <p>Prevents others from seeing all participants' video contribution (mute).</p>  <p>Allows others to see all participants' video contribution (stop muting).</p>	<p>If no participants are currently muted, the <i>stop muting</i> controls will be unavailable. If all participants are currently muted, the <i>mute</i> controls will be unavailable.</p> <p>The presence of these controls is controlled by the <b>Participant list controls</b> setting on the <b>Settings &gt; User interface</b> page (refer to <a href="#">Customizing the user interface</a>).</p>
<b>Disconnect</b>	 <p>Disconnects all participants from the conference.</p>	If this conference was created ad hoc, then disconnecting all the participants will terminate the conference as well.

Field	Field description	Usage tips
<b>View</b>	 Selects <i>voice-switched</i> view for all participants (see <a href="#">Customizing a participant's layout view</a> ).  Selects full screen view for all participants.  Displays the layout selection panel from which you can select a layout view that all current participants will see. This panel offers the same choices as going to <b>Conferences &gt; Custom layout</b> . (See <a href="#">Customizing layout views</a> ).	If all participants are watching the voice-switched view, then this control will be unavailable.
<b>Control</b>	 Prevents participants from changing their own view layout from their endpoint.  Allows participants to change their own view layout from their endpoint.	If no participants may control their own view layout, the <i>prevent</i> control will be unavailable. If all participants may control their own layout, the <i>allow</i> control will be unavailable.

## Previous participants

Refer to the table below to get details about the previous participant list. To delete this list, click **Clear previous participants record**.

Field	Field description	Usage tips
<b>Type</b>	Indicates whether the participant is an H.323, SIP, or VNC endpoint.	
<b>Participant</b>	Displays the name of the participant (endpoint).	
<b>Controls</b>	 Re-connects a participant to the conference.	<p>A previous participant can only be re-connected to a conference if their endpoint is a pre-configured endpoint.</p> <p>Note that if the original call to the conference was from the endpoint rather than the endpoint being invited by the MCU, then for the MCU to recognize it, that endpoint must be configured with <b>Call-in match parameters</b>. If the MCU did not recognize that the call was from a pre-configured endpoint then the re-connect control will not be available.</p> <p>For more information about configuring endpoints on the MCU refer to <a href="#">Configuring</a></p>

Field	Field description	Usage tips
		<a href="#">H.323 endpoints</a> and <a href="#">Configuring SIP endpoints</a> .
<b>Status</b>	Displays the time at which a participant disconnected from the conference and who initiated the disconnect.	

## Pre-configured participants

Refer to the table below to get details about pre-configured endpoints.

Field	Field description	Usage tips
<b>Type</b>	Indicates whether the participant is an H.323, SIP, or VNC endpoint.	
<b>Name</b>	Displays the name of the endpoint.	
<b>Status</b>	Displays the connection status of the pre-configured participant.	This field shows which participants have connected, which have disconnected, and reasons for failure if a participant has failed to connect to the conference.

## Summary information

You may wish to inform participants about conference details such as start time, and so on. Click the **Summary information** icon to display further details about a conference. This information may be copied to the clipboard for convenience.

## Streaming the conference

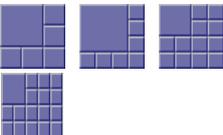
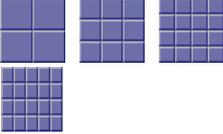
Click this icon to stream the conference. This is exactly the same as clicking the **Streaming** link in the Conference list. Both take you to the conference streaming page.

This option is only available if you have streaming enabled on your MCU. To enable streaming, go to **Settings > Streaming**. See [Using streaming to view conferences](#) for more information.

## Customizing a participant's layout view

Every person viewing a conference (whether they are contributing video of their own or using streaming to watch other participants) sees a conference layout view. This view divides the video screen into a collection of panes with participants' video streams displayed in those panes. You can customize this view by choosing this option:

1. Go to **Conferences**.
2. Click a Conference name and then click on a participant's name.
3. Click the **Display** tab.

Field	Field description	Usage tips
<b>Preview</b>		
<b>Video display</b>	Displays static sample of video screen in the currently selected layout.	Click on the image to refresh the picture
<b>Border</b>	Adds a border thickness to display around the video image.	If the image is displaying off the edges of the participant's screen, add a border until the image displays properly.
<b>Layout</b>		
<b>Family 1</b>	Give prominence to one participant over the others.	The number of contributing conference participants determines the size of the large pane.
		
<b>Family 2</b>	Displays a single participant.	
		
<b>Family 3</b>	Displays the four most active participants without seeing them scaled down to a small size if there are lots of other participants.	Used when there are five or more video participants.
		
<b>Family 4</b>	Gives equal prominence to up to 20 conference contributors, and is useful for a "roll call" of active participants.	The MCU automatically changes the layout to the 3 x 3 arrangement, and will continue to use this layout for up to 9 participants. With 10 or more participants, the 4 x 4 view is used, and with 17 or more participants the 5 x 4 view will be used. The MCU will then continue to use this layout even if there are more than 20 participants.
		

Field	Field description	Usage tips
<b>Family 5</b>  	<p>Gives prominence to two participants in the center of the view while showing smaller versions of other participants' video streams above and below.</p>	<p>This view is useful for observing a dialog between two participants or for viewing slides and a presenter.</p>
<b>Conference custom layout</b>	<p>Click <b>Edit</b> to choose or create a custom layout to be used by any participant.</p>	<p>See <a href="#">Customizing layout views</a>.</p>
<b>Participant custom layout</b>	<p>Click <b>Edit</b> to choose a pre-configured custom layout for this participant.</p>	<p>See <a href="#">Selecting a custom participant layout</a>.</p>
<b>Controls</b>		
	<p>Prevents this participant from changing their own conference layout using the far-end camera controls on their video endpoint.</p>	<p>A participant can still change their layout view using the web interface.</p>
	<p>Allows this participant to control their own conference view.</p>	
	<p>Send "normal" format video to this participant - this means that the video is intended to be displayed on a screen whose dimensions (width and height) are in the ratio 4:3.</p>	<p>The available options for the resolution of transmitted video are controlled by the <b>Transmitted video resolutions</b> setting which can be configured for the whole unit on the <b>Settings &gt; Conferences</b> page or per configured endpoint (refer to <a href="#">Configuring an H.323 endpoint</a> and/or <a href="#">Configuring a SIP endpoint</a>).</p>
	<p>Send "widescreen" format video to this participant - this means that the video is intended to be displayed on a screen whose dimensions (width and height) are in the ratio 16:9.</p>	
	<p>Do not allow this participant to start contributing content channel video.</p>	<p>For more information, see <a href="#">Content per-participant parameters</a>.</p>
	<p>Allow this participant to start contributing content channel video.</p>	
<b>Focused participant</b>		
	<p>Indicates which participant appears in the largest pane.</p>	<p>Depending on the types of participant in the conference (i.e. content channel, endpoint, audio-only participant), you can select from these options:</p> <ul style="list-style-type: none"> <li>▶ <i>Voice-switched</i>: the loudest participant will given the largest pane in the conference (the default setting)</li> <li>▶ <i>Content channel</i>: the content channel will be given the largest pane in the conference</li> <li>▶ <i>&lt;participant name&gt;</i>: the participant that you select will be given the largest pane in the conference</li> </ul>

Field	Field description	Usage tips
		This setting will be overridden if a participant is identified as "important" on the Conference Participants List (see <a href="#">Viewing the conference participants list</a> ).

## Controlling the near-end camera

You can control the viewing angle, zoom, and focus of the camera on a participant's video endpoint. You can customize this view by choosing this option:

1. Go to **Conferences**.
2. Click a Conference name and then click on a participant's name.
3. Click the **Camera** tab.

Field	Field description	Usage tips
<b>Movement</b>	Click on one of the directional arrows to change the view direction of the camera.	Not all endpoints will respond to these controls. In particular, endpoints with fixed cameras such as most webcams.
<b>Zoom</b>	Click on one of the magnifying options to zoom the view in or out.	Not all endpoints will respond to the zoom controls, possibly including those that do respond to the <b>Movement</b> controls
<b>Focus</b>	Click on one of the adjustment options to improve the clarity of the image.	Not all endpoints will respond to the focus controls, possibly including those that do respond to the <b>Movement</b> controls

## Video mute

You may not want a participant to be visible in a conference, for example if they are only supervising, but not contributing. The MCU allows this. Refer to the table below for details of the controls available:

Field	Field description	Usage tips
	Prevents others from seeing this participant's video in conference views (mute).	This setting applies only to conference views. The participant's video will still appear in the conference participant list (see <a href="#">Viewing conference participants list</a> ), and in the auto attendant if a conference is not PIN-protected (see <a href="#">Using an auto attendant</a> ).
	Allows others to see this participant's video in conference views (stop muting)	

## Viewing a participant's audio signals

In general, the audio settings are for advanced troubleshooting. You can view these settings by selecting this option:

1. Go to **Conferences**.
2. Click a Conference name and then click on a participant's name.
3. Click the **Audio** tab.

These settings can help you resolve audio problems in conferences. For example:

- ▶ Waveforms display a snapshot of the audio waveform received from the participant's endpoint. If the participant is silent or speaking very quietly, the preview displays a horizontal green line. If the participant is speaking, the line is wavy.
- ▶ **AGC control**: use this setting if participants are too loud or too quiet. For example, if a participant is shouting on the remote end, but is barely audible to others in the conference.
- ▶ **Apply fixed gain**: a volume control which raises the decibel level by the specified amount. Enter a positive value to raise the volume of the participant and a negative value to reduce it.
- ▶ **Mute in-band DTMF tones**: a control to prevent DTMF tones being transmitted to other participants. The MCU will still detect and act on the DTMF tones; this control prevents the tones from being audible to participants.

### Audio mute

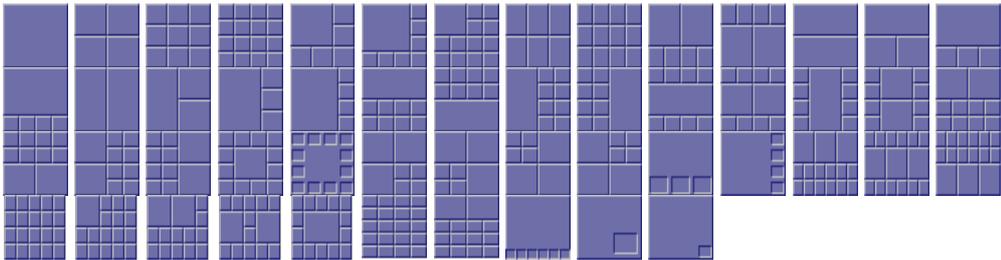
You can mute a participant's audio. Refer to the table below for details of the controls available.

Field	Field description	Usage tips
	Prevents others from hearing this participant (mute).	
	Allows others to hear this participant's audio (stop muting).	

## Selecting a custom participant view

You can add a custom layout to the choice of layouts available for this participant. You can customize this view by choosing this option:

1. Go to **Conferences**.
2. Click a Conference name and then click on a participant's name.
3. Click the **Custom layout** tab.

Field	Field description	Usage tips
<b>Current status</b>	Indicates whether a custom layout can be used for this participant.	When you click a new layout from the <b>Available layouts</b> , <b>Enabled</b> is automatically chosen with the most recent layout selected.
<b>Available layouts</b>	Displays all pre-configured layouts that are available to the participant. Click a layout to select it and make it available for the participant.  These layouts are designed to be suitable for all video conferencing situations:    These layouts are designed to be suitable for displaying composed views of standard (4:3) video streams on widescreen (16:9) displays:  	You can only select one custom layout per participant.

Note that the layouts depicted with slightly darker shading for the 'picture in picture' participants indicate those layouts where the non-focused participants are slightly transparent and the focused participant can be seen 'through' the non-focused participants. These layouts are:



and

# Displaying statistics for a participant

You can view statistics about the video and audio streams between individual participants (endpoints) and the MCU by choosing this option:

1. Go to **Conferences**.
2. Click a Conference name and then click on a participant's name.
3. Click the **Statistics** tab.

If the participant is participating in the conference using audio only, the values for the video settings are not populated.

## Media statistics

Media statistics provide detailed information about the actual voice and video streams (Realtime Transport Protocol (RTP) packets).

When examining media statistics for a conference's [streaming viewers](#), there are many fewer applicable values and so a different, more limited, set of information is displayed. In this case, the MCU shows how many streams of each outgoing audio and video format are being transmitted.

Refer to the table below for additional information.

Field	Field description	Usage tips
<b>Audio</b>		
<b>Receive stream</b>	The audio codec in use, along with the current packet size (in milliseconds) if known.	If the MCU has received information that an endpoint has been muted at the far end, this will be indicated here.
<b>Receive address</b>	The IP address and port from which the media is originating.	
<b>Encryption</b>	Whether or not encryption is being used on the audio receive stream by this endpoint.	This field will only appear if the encryption feature key is present on the MCU.
<b>Received jitter</b>	The apparent variation in arrival time from that expected for the media packets (in milliseconds). The current jitter buffer also displays in parentheses.	<p>You should expect to see small values for this setting. Consistently large numbers typically imply potential network problems.</p> <p>The jitter buffer shows the current playout delay added to the media to accommodate the packet arrival jitter. Large jitter values indicate a longer buffer.</p>
<b>Received energy</b>	Represents the audio volume originating from the endpoint.	
<b>Packets received</b>	The number of audio packets destined for the MCU from this endpoint.	
<b>Packet errors</b>	The number of packet errors, including sequence errors, and packets of the wrong type.	You should expect to see small values for this setting. Consistently large numbers typically imply potential network

Field	Field description	Usage tips
		problems.
<b>Frame errors</b>	Frame errors, as $A/B$ where $A$ is the number of frame errors, and $B$ is the total number of frames received.	A frame is a unit of audio, the size of which is dependent on codec.  You should expect to see small values for this setting. Consistently large numbers typically imply potential network problems.
<b>Media information</b>	If the time stamps or marker bits (or both) are detected to be unreliable in the incoming video stream, information will be displayed here.	This field is not displayed when there is no problem with the time stamps and marker bits. Where there is a problem the following text is displayed: "Media timestamps unreliable", "Media marker bits unreliable", or both if both conditions detected.
<b>Transmit stream</b>	The audio codec being sent from the MCU to the endpoint, along with the chosen packet size in milliseconds.	
<b>Transmit address</b>	The IP address and port to which the media is being sent.	
<b>Encryption</b>	Whether or not encryption is being used on the audio transmit stream by this endpoint.	This field will only appear if the encryption key is present on the MCU.
<b>Packets sent</b>	A count of the number of packets that have been sent from the MCU to the endpoint.	
<b>RTCP receive address</b>	The IP address and port to which RTCP (Real Time Control Protocol) packets are being sent for the audio and video streams.	
<b>Receiver reports</b>	A count of the number of "receiver report" type RTCP packets seen by the MCU.	A single RTCP packet may contain more than one report of more than one type. These are generally sent by any device receiving RTP (Real Time Protocol) media from the network and are used for auditing bandwidth, errors, and so on by the MCU.
<b>Packet loss reported</b>	A count of the reported packet loss on the control channel.	
<b>Sender reports</b>	A count of the number of "sender report" type RTCP packets sent by the MCU.	These are typically sent by any device that is sending RTP media.
<b>RTCP transmit address</b>	The IP address and port to which the MCU is sending RTCP packets about this stream.	
<b>Receiver reports</b>	A count of the number of "receiver report" type RTCP packets seen by the MCU.	A single RTCP packet may contain more than one report of more than one type. These are generally sent by any device receiving RTP (Real Time Protocol) media from the network and are used for auditing bandwidth, errors, and so on by the MCU.

Field	Field description	Usage tips
<b>Sender reports</b>	A count of the number of "sender report" type RTCP packets received by the MCU.	These are typically sent by any device that is sending RTP media.
<b>Packets sent</b>	A count of the number of packets that have been sent from the MCU to the endpoint.	
<b>Video (primary channel and content shown separately)</b>		
<b>Receive stream</b>	The codec in use and the size of the picture that the MCU is receiving from the specific participant. If the picture is a standard size (for example, CIF, QCIF, 4CIF, SIF) then this name is shown in parentheses afterwards.	
<b>Receive address</b>	The IP address and port (<IP address>:<port>) of the device from which video is being sent	
<b>Encryption</b>	Whether or not encryption is being used on the video receive stream from this endpoint.	This field will only appear if the encryption key is present on the MCU.
<b>Channel bit rate</b>	The negotiated bit rate available for the endpoint to send video in.	This value represents the maximum amount of video traffic that the remote endpoint will send to the MCU. It may send less data than this (if it does not need to use the full channel bit rate or the MCU has requested a lower rate), but it should not send more.
<b>Receive bit rate</b>	The bit rate (in bits per second) that the MCU has requested that the remote endpoint sends. The most-recently measured actual bit rate displays in parentheses.	<p>This value might be less than the <b>Channel bit rate</b> if:</p> <ul style="list-style-type: none"> <li>▶ the MCU detects that the network path to the remote endpoint has insufficient capacity to maintain a higher traffic rate</li> <li>▶ that endpoint's video stream's position in the active conference compositions does not require it</li> <li>▶ it has been necessary to reduce the video bit rate because of the overall call bit rate; the audio bit rate plus the video bit rate should not exceed the call bit rate</li> </ul> <p>For example, if all participants in the conference were watching a single participant at full screen, no other participants' video streams would be needed at all. So the MCU would request that those streams were sent at a low bit rate in order to avoid needless use of network bandwidth.</p> <p>If the receive bit rate has been limited to below the maximum channel bit rate, the</p>

Field	Field description	Usage tips
		reason for this limitation can be seen by moving over the  icon.
<b>Received jitter</b>	Represents the variation in video packet at arrival time at the MCU.	
<b>Offset after audio</b>	The number of milliseconds by which the video follows the audio. Some endpoints send audio and video out-of-sync. The MCU ensures audio and video are played together.	
<b>Packets received</b>	The number of video packets destined for the MCU from this endpoint	
<b>Packet errors</b>	Video packet-level errors such as sequence discontinuities, incorrect RTP details, and so on. This is not the same as packets where the content (the actual video data) is somehow in error.	This value does not represent packets in which the actual video data in the packets is in error.
<b>Frame rate</b>	The frame rate of the video stream currently being received from the endpoint.	
<b>Frame errors</b>	The number of frames with errors versus the total number of video frames received.	
<b>Transmit stream</b>	The codec, size and type of video being sent from the MCU to the endpoint.	
<b>Transmit address</b>	The IP address and port of the device to which the MCU is sending video.	
<b>Encryption</b>	Whether or not encryption is being used on the video transmit stream to this endpoint.	This field will only appear if the encryption key is present on the MCU.
<b>Channel bit rate</b>	The negotiated available bandwidth for the MCU to send video to the endpoint in.	
<b>Transmit bit rate</b>	The bit rate the MCU is attempting to send at this moment, which may be less than the channel bit rate which is an effective maximum. The actual bit rate, which is simply the measured rate of video data leaving the MCU, displays in parentheses.	<p>The <b>Transmit bit rate</b> value might be less than the <b>Channel bit rate</b> if :</p> <ul style="list-style-type: none"> <li>▶ the remote endpoint receiving the video stream from the MCU has sent flow control commands to reduce the bit rate</li> <li>▶ it has been necessary to reduce the primary video bit rate to allow sufficient bandwidth for a content video stream</li> </ul> <p>If the transmit bit rate has been limited to below the maximum channel bit rate, the reason for this limitation can be seen by moving over the  icon.</p>
<b>Packets sent</b>	The number of video packets sent from the MCU to this endpoint.	

Field	Field description	Usage tips
<b>Frame rate</b>	The frame rate of the video stream currently being sent to the endpoint.	
<b>Temporal/spatial</b>	A number that represents the tradeoff between video quality and frame rate.	A smaller number implies that the MCU prioritizes sending quality video at the expense of a lower frame rate. A larger number implies that the MCU is prepared to send lower quality video at a higher frame rate.

## Control statistics

Control statistics provide information about the control channels that are established in order that the endpoints can exchange information about the voice and video streams (Real Time Control Protocol (RTCP) packets). Refer to the table below for additional information.

Field	Field description	Usage tips
<b>Audio</b>		
<b>RTCP receive address</b>	The IP address and port to which RTCP (Real Time Control Protocol) packets are being received for the audio and video streams	
<b>Receiver reports</b>	A count of the number of "receiver report" type RTCP packets seen by the MCU.	A single RTCP packet may contain more than one report of more than one type. These are generally sent by any device receiving RTP (Real Time Protocol) media from the network and are used for auditing bandwidth, errors, and so on by the MCU.
<b>Packet loss reported</b>	Media packet loss reported by receiver reports sent to the MCU by the far end.	
<b>Sender reports</b>	A count of the number of "sender report" type RTCP packets received by the MCU.	These are typically sent by any device that is sending RTP media.
<b>Other</b>	A count of the number of reports seen by the MCU that are neither sender nor receiver reports.	
<b>RTCP transmit address</b>	The IP address and port to which the MCU is sending RTCP packets about this stream.	
<b>Packets sent</b>	The number of packets sent.	
<b>Video (primary channel and content shown separately)</b>		
<b>RTCP receive address</b>	The IP address and port to which RTCP (Real Time Control Protocol) packets are being sent for the audio and video streams.	
<b>Receiver reports</b>	A count of the number of "receiver report" type RTCP packets seen by the MCU.	A single RTCP packet may contain more than one report of more than one type. These are generally sent by any device receiving RTP (Real Time Protocol) media

Field	Field description	Usage tips
		from the network and are used for auditing bandwidth, errors, and so on by the MCU.
<b>Packet loss reported</b>	A count of the reported packet loss on the control channel.	
<b>Sender reports</b>	A count of the number of "sender report" type RTCP packets sent by the MCU.	These are typically sent by any device that is sending RTP media.
<b>Other</b>	A count of the number of reports seen by the MCU that are neither sender nor receiver reports.	
<b>RTCP transmit address</b>	The IP address and port to which the MCU is sending RTCP packets about this stream.	
<b>Packets sent</b>	The number of packets sent.	
<b>Fast update requests</b>	The number of fast update requests sent and received.	
<b>Flow control messages</b>	The number of flow control messages sent and received.	

## Sending messages to individual participants

You can send a short text message to a specific participant currently in the conference. To do this:

1. Go to **Conferences**.
2. Click a Conference name and then click on a participant's name.
3. Click the **Send message** tab.

This message appears overlaid on the participant's view, even if the participant is viewing the conference via streaming.

Field	Field description	Usage tips
<b>Message text</b>	Enter the message to send to this participant.	<p>Messages must be fewer than 256 characters, but depending on the viewing screen, messages at the higher-end of this limit might not display properly. So, consider limiting messages to about 180 characters. Messages longer than 256 characters will not be truncated; they will not display at all.</p> <p>You can disable this setting from <b>Settings &gt; Conferences</b> (see <a href="#">Conference settings</a>).</p>
<b>Position</b>	The vertical position of the message on the conference display.	Choose from the top, middle, or bottom of the conference display.
<b>Duration</b>	Indicates how long the message appears on the endpoint's video screen.	The default setting is 30 seconds. To remove a message before it times out, click <b>Clear message</b> .

## Sending DTMF to an audio bridge

If you are cascading a conference on the MCU to an audio conferencing bridge, you might want to send DTMF tones to that audio bridge for the purpose of entering a conference ID number and PIN. To access the DTMF keypad:

1. Go to **Conferences** and click on the conference name.
2. From the list of participants, click on the audio bridge's name.
3. Click the **Send DTMF** tab.

Pressing the keys on the DTMF keypad causes the MCU to generate the DTMF tones and send them in-band in the audio channel to the participant (in this case, the audio conferencing bridge). In this way, you can dial the audio conferencing bridge into a conference and then listen in to hear the instructions from the audio bridge. When the audio bridge requires DTMF entries, you can enter the required numbers by using the DTMF keypad.

Note that if you know exactly what the audio conferencing bridge is going to require when it is dialed into a conference, you can configure the bridge as an endpoint and set a DTMF sequence to be sent to that bridge. For more information about configuring endpoints refer to [Configuring H.323 endpoints](#) and/or [Configuring SIP endpoints](#).

# Displaying diagnostics for a participant

You can view diagnostics for an individual participant's connection to the MCU by choosing this option:

1. Go to **Conferences**.
2. Click a conference name and then click on a participant's name.
3. Click the **Diagnostics** tab.

## Participant diagnostics

This page shows various low-level details pertaining to the endpoint's communication with the MCU. You are not likely to need to use any of the information on this page except when troubleshooting specific issues under the guidance of customer support.

## Moving a participant

You can move participants between conferences. Participants can be moved to any conference, but although participants can be moved from auto attendants they cannot be moved *to* them.

To move a participant:

1. Go to **Conferences > Move participants**.
2. Click and drag the name of the participant to the required destination conference.

Note that participants dragged and dropped into a conference on the web interface will be chairperson participants. (For more information about chairperson and guest participants, refer to [Adding and updating conferences](#).)

To move multiple participants:

1. Go to **Conferences > Move participants**.
2. Select the check boxes for every participant you want to move and drag them to the required destination conference.

To disconnect a participant from a conference:

1. Go to **Conferences > Move participants**.
2. Click and drag the name of the participant in to **Remove participant** area at the bottom of that page.

Note that dragging and dropping a participant in to the **Remove participant** area simply disconnects the participant from the conference. It does not remove them from a scheduled conference. That is, when the conference next runs that participant will be called if their endpoint is a pre-configured endpoint for that conference.

## Creating auto attendants

The Cisco TelePresence MCU allows you to configure auto attendants on it, which allows users to more easily join conferences.

- ▶ [Displaying the auto attendant list](#)
- ▶ [Adding and updating an auto attendant](#)
- ▶ [Adding a custom banner](#)

# Displaying the auto attendant list

You can display an overview of the configured auto attendants on the MCU:

1. Go to **Conferences**.
2. Click the **Auto attendants** tab.

Field	Field description	Usage tips
<b>Name</b>	The name of the auto attendant	
<b>Numeric ID</b>	The number that you can dial to connect to the auto attendant	
<b>H.323 gatekeeper</b>	<p>The status of an auto attendant with respect to its gatekeeper registration. The possible states are:</p> <ul style="list-style-type: none"> <li>▶ <i>n/a</i> This auto attendant is not configured to be registered with the gatekeeper; because of this, there is no applicable registration status to show.</li> <li>▶ <i>Registering</i> This auto attendant is in the process of registering with the gatekeeper.</li> <li>▶ <i>Deregistering</i> The auto attendant is in the process of unregistering with the gatekeeper. This might occur if: <ul style="list-style-type: none"> <li>• Gatekeeper registration has been turned off (either for that auto attendant only or for the entire MCU)</li> <li>• The configured gatekeeper has just been changed and the MCU is in the process of unregistering from the previous gatekeeper before registering its auto attendants with the new one.</li> </ul> </li> <li>▶ <i>Re-registration pending / Retry timeout</i> If the MCU fails to register an auto attendant with the gatekeeper, it enters these states temporarily before re-attempting the registration.</li> <li>▶ <i>&lt;number&gt; registered</i> The auto attendant has been registered successfully with the gatekeeper and can be contacted using the number indicated.</li> </ul>	For tips on configuring gatekeepers, see <a href="#">H.323 g atekeeper settings</a> .
<b>SIP registrar</b>	The status of an auto attendant with respect to its SIP registration. The possible	For tips on configuring SIP, see <a href="#">SIP settings</a> .

Field	Field description	Usage tips
	<p>states are:</p> <ul style="list-style-type: none"> <li>▶ <i>n/a</i> This auto attendant is not configured to be registered with the SIP registrar; because of this, there is no applicable registration status to show.</li> <li>▶ <i>Registering</i> This auto attendant is in the process of registering with the SIP registrar.</li> <li>▶ <i>Deregistering</i> The auto attendant is in the process of unregistering with the SIP registrar. This might occur if: <ul style="list-style-type: none"> <li>• SIP registration has been turned off (either for that auto attendant only or for the entire MCU)</li> <li>• The configured SIP registrar has just been changed and the MCU is in the process of unregistering from the previous SIP registrar before registering its auto attendants with the new one.</li> </ul> </li> <li>▶ <i>Re-registration pending / Retry timeout</i> If the MCU fails to register an auto attendant with the SIP registrar, it enters these states temporarily before re-attempting the registration.</li> <li>▶ <i>&lt;number&gt; registered</i> The auto attendant has been registered successfully with the SIP registrar and can be contacted using the number indicated.</li> </ul>	
<b>Security</b>	Whether a PIN has been configured to restrict access to the auto attendant	
<b>Calls</b>	The total number of calls received by the auto attendant since the last restart	
<b>Banner</b>	A thumbnail of the custom banner, if one has been specified	

# Adding and updating an auto attendant

Auto attendants simplify the way participants can join conferences. By calling an auto attendant using their video endpoint, a participant can choose from menu options and join or start conferences. No gateway or gatekeeper is required. (The auto attendant is configured on the MCU.)

## Adding an auto attendant

To add an auto attendant:

1. Go to **Conferences**.
2. Click the Auto attendants tab.
3. Click **Add new auto attendant**.
4. Refer to the table below for the most appropriate settings for the auto attendant.
5. After entering the settings, click **Add auto attendant**.

## Updating an auto attendant

To update an existing auto attendant:

1. Go to **Conferences**.
2. Click the Auto attendants tab.
3. Click the name of an auto attendant.
4. Refer to the table below for the settings to change for this auto attendant.
5. After updating the settings, click **Update auto attendant**.

Field	Field description	Usage tips
<b>Auto attendant</b>		
<b>Name</b>	The name of the auto attendant.	
<b>Numeric ID</b>	The number with which to register the auto attendant on the gatekeeper and/or SIP registrar.	See <a href="#">Configuring H.323 gatekeeper settings</a> and <a href="#">SIP settings</a> for details.
<b>Numeric ID registration</b>	Select <i>H.323 gatekeeper</i> if you want the <b>Numeric ID</b> to be registered with the H.323 gatekeeper; select <i>SIP registrar</i> if you want the <b>Numeric ID</b> registered with the SIP registrar.	Note that for SIP, you must configure the ID with the SIP registrar for the MCU to be able to register that ID.
<b>PIN</b>	Assigns a password to the auto attendant.	If you set a PIN, all participants using the auto attendant will be required to enter this password.
<b>Re-enter PIN</b>	Verifies the password.	
<b>Creation of new conferences</b>	If Enabled, displays the <i>Create new conference</i> option on the auto attendant so that the participant can create new ad hoc conferences using the auto attendant.	If disabled, participants will not be able to create new conferences from the auto attendant.  Note that this option will not be available if you have enabled Media port reservation

Field	Field description	Usage tips
		on the <b>Settings &gt; Conferences</b> page. When using port reservation mode, there can be no ad hoc conferences. For more information about port reservation, refer to <a href="#">Reservation of MCU media ports</a> .
<b>Access to ad hoc conferences</b>	If Enabled, displays ad hoc conferences as well as scheduled conferences as options on the auto attendant.	If disabled, participants can only view scheduled conferences; ad hoc conferences will not be shown.  Note that this option will not be available if you have enabled Media port reservation on the <b>Settings &gt; Conferences</b> page. When using port reservation mode, there can be no ad hoc conferences. For more information about port reservation, refer to <a href="#">Reservation of MCU media ports</a> .
<b>Accessible conferences</b>		
<b>All scheduled conferences</b>	Enables all auto attendant participants to join any conferences scheduled to start while they are using the auto attendant.	If you select <b>All scheduled conferences</b> , this does not include conferences configured as private.
<b>Selected scheduled conferences</b>	Choose the conferences to list on the auto attendant by selecting check boxes.	Note that even conferences that have been configured as private conferences will appear on this list. So as the admin user, you can enable private conferences to appear on an auto attendant. If you leave the conference unselected, it will not appear on the auto attendant.
<b>Links to other auto attendants</b>		
<b>Select the name of any other configured auto attendant that you want to be accessible from the auto attendant that you are adding.</b>		

## Adding a custom auto attendant banner

You can add a custom banner image to any auto attendant configured on the MCU as follows:

1. Go to **Conferences**.
2. Click the **Auto attendants** tab.
3. Click the name of a configured auto attendant.
4. Click the **Banner** tab.
5. Refer to the table below to determine the most appropriate settings.

Field	Field description	Usage tips
<b>Auto attendant banner</b>		
<b>Default</b>	Chooses the default Cisco TelePresence MCU graphic to use for your banner.	
<b>Specific to this auto attendant</b>	The custom banner identified for this auto attendant. Click <b>Remove banner</b> to remove this graphic as the banner. Click <b>Update</b> after uploading a new graphic.	Nothing displays here until you upload the custom graphic as described below.
<b>Banner upload</b>		
<b>Banner for this auto attendant</b>	The custom graphic to be used for a banner. Click <b>Browse</b> to locate the file on your hard drive.	The image file can be a JPEG, GIF or Windows BMP format with a maximum size of 352 x 64 pixels.
<b>Background color</b>	Sets a custom background color. Enter the color values in each field. Click <b>Upload new file</b> to display.	The background color is used if the banner is smaller than 352 x 64 pixels.

## Displaying the endpoint list

To display the Endpoint List, go to **Endpoints**.

The Endpoint List displays all endpoints that have been configured within the MCU.

To add a new H.323 endpoint, select **Add H.323**.

To add a new SIP endpoint, select **Add SIP**.

To add a new VNC endpoint, select **Add VNC**.

To delete configured endpoints, check the ones you want to delete and select **Delete selected**.

Field	Field description
<b>Name</b>	The name of the endpoint.
<b>Address</b>	The IP address, host name, H.323 ID, E.164 number, or SIP URI of the endpoint.
<b>Type</b>	Whether it is an H.323, SIP, or VNC endpoint.

## Configuring H.323 endpoints

You can configure H.323 endpoints to work with the MCU by choosing **Endpoints > Add H.323**. This makes it easier to add endpoints to conferences because you can select names from a list rather than adding network addresses.

A Cisco TelePresence IP VCR can be configured as an H.323 endpoint and added as a participant in a conference. If the IP VCR is configured to do so, it will start recording as soon as the conference starts. You can also configure a folder's Recording ID as an endpoint and in this way, when a conference starts, the IP VCR can start recording directly into a specific folder. For more information about using the IP VCR in this way, refer to the IP VCR's online help.

Recordings on a Cisco TelePresence IP VCR can be configured as H.323 endpoints. In this way, an audio-only participant can contribute an IP VCR recording as his video stream (using the associated endpoint function, see [Adding and updating users](#)). This function is also useful where you have a recording that you might like to view within a conference.

Refer to the table below for tips on adding an H.323 endpoint to the MCU. After entering the settings, click **Add endpoint**.

Field	Field description	Usage tips
<b>Name</b>	The name of the endpoint.	
<b>Address</b>	The IP address, host name, or an E.164 address (phone number).	You can configure this endpoint as needing to be reached via an H.323 gateway without that gateway being already configured on the MCU. To do this, set this field to be <code>&lt;gateway address&gt;!&lt;E.164&gt;</code> .
<b>H.323 gateway</b>	The gateway through which the endpoint connects.	To configure a gateway on the MCU, go to <b>Gateways</b> . For more information, refer to <a href="#">Adding and updating gateways</a> .
<b>DTMF sequence</b>	The DTMF sequence to be sent to an audio conferencing bridge.	<p>Allows the MCU to send DTMF tones to an audio bridge after the audio bridge has answered the call. In this way, the MCU can navigate audio menus. This is useful where a conference on the MCU dials out to an audio-only conference on an audio bridge.</p> <p>You can configure an audio bridge with a DTMF sequence as a pre-configured endpoint (either H.323 or SIP) which can then be added to any conference. Alternatively, you can add the audio bridge to an individual conference as a participant. You must specify the DTMF sequence in the Call-out parameters for that endpoint. The DTMF sequence can include digits 0-9 and * and #. There is a two second pause after the call connects after which the MCU will send the DTMF</p>

Field	Field description	Usage tips
		<p>tones which are sent one every half second. You can insert as many additional two second pauses as you want by inserting commas into the DTMF sequence. Each comma represents a two second pause.</p> <p>For example, you want the MCU to dial out to a PIN-protected audio conference on an audio bridge. The conference ID is 555 and the PIN is 888. The audio bridge requires that you press # after entering the ID and after entering the PIN. In this example the DTMF sequence is: 555#,888#. The two commas represent a four second pause which allows the audio bridge's automated menu system time to process the ID and request the PIN.</p>
<b>Call-in match parameters</b>	<p>These fields are used to identify incoming calls as being from the endpoint:</p> <ul style="list-style-type: none"> <li>▶ <b>Name:</b> This must be the name that the endpoint sends to the MCU</li> <li>▶ <b>IP address:</b> The IP address of the endpoint</li> <li>▶ <b>E.164:</b> The E.164 number with which the endpoint is registered with the gatekeeper</li> </ul>	<p>The endpoint is recognized if all filled-in fields in this section are matched. Fields left blank are not considered in the match.</p> <p>When you configure <b>Call-in match parameters</b>, an endpoint will be recognized as <i>this</i> pre-configured endpoint and the <b>Conferencing parameters</b> will be applied to a call from this endpoint.</p>
<b>Display name override</b>	<p>The name that will be displayed in a conference as a label for this endpoint.</p>	<p>The name you enter here will override any default name configured on the endpoint. It will also override any other default name that might appear for an endpoint. For example, an endpoint's default name can be the name of the gateway through which the call was placed, or if the endpoint is called-in via a gatekeeper, its E.164 number.</p> <p>Note that once an endpoint has connected, you cannot change the display name.</p>
<b>Motion / sharpness trade off</b>	<p>Select whether to use the unit-wide setting for motion/sharpness trade off, or configure an individual setting for use with this endpoint. Select from:</p> <ul style="list-style-type: none"> <li>▶ <i>&lt;use box-wide setting&gt;</i>: this is the default value. In this case, the connection to the endpoint will use the motion / sharpness tradeoff setting from the <b>Settings &gt; Conferences</b> page.</li> <li>▶ <i>Favor motion</i>: the MCU will try and use a high frame rate. That is, the</li> </ul>	<p>The settings for motion (frames per second) and sharpness (frame size or resolution) are negotiated between the endpoint and the MCU. This setting controls how the MCU will negotiate the settings to be used with this endpoint.</p>

Field	Field description	Usage tips
	<p>MCU will strongly favor a resolution of at least 25 frames per second</p> <ul style="list-style-type: none"> <li>▶ <i>Favor sharpness</i>: the MCU will use the highest resolution that is appropriate for what is being viewed</li> <li>▶ <i>Balanced</i>: the MCU will select settings that balance resolution and frame rate (where the frame rate will not be less than 12 frames per second)</li> </ul>	
<b>Transmitted video resolutions</b>	<p>Select the setting for transmitted video resolutions from the MCU to this endpoint. This setting overrides the unit-wide setting on the <b>Settings &gt; Conferences</b> page</p>	<p>Retain the default setting (<i>use box-wide setting</i>) unless you are experiencing problems with the display of certain resolutions by this endpoint.</p> <p>Endpoints advertise the resolutions that they are able to display. The MCU then chooses from those advertised resolutions, the resolution that it will use to transmit video. However, some endpoints do not display widescreen resolutions optimally. Therefore, you might want to use this setting to restrict the resolutions available to the MCU for transmissions to this endpoint.</p>
<b>Content contribution</b>	<p>Whether this endpoint is permitted to contribute the conference content channel. Select from:</p> <p>&lt;use conference default&gt;: this endpoint will use the <b>Content contribution from endpoints</b> setting from the per-conference configuration.</p> <p><i>Enabled</i>: This endpoint is allowed to contribute the content channel, even if content contribution from endpoints is disabled in the per-conference configuration.</p> <p><i>Disabled</i>: This endpoint is not allowed to contribute the conference channel, even if content contribution from endpoints is enabled in the per-conference configuration.</p>	<p>This setting is provided to allow you to individually configure whether or not an endpoint is allowed to contribute content to a conference.</p> <p>To use the content channel, the <b>Content status</b> must be enabled at the unit-wide level (on the <a href="#">Settings &gt; Content</a> page) and for any given conference <b>Content channel video</b> must also be enabled in the <a href="#">per-conference configuration</a>.</p>
<b>Content receive</b>	<p>Whether this endpoint is allowed to receive a separate content stream when in a conference.</p>	<p>This setting is provided to allow you to individually configure whether or not an endpoint is allowed to receive content from a conference.</p> <p>To use the content channel, the <b>Content status</b> must be enabled at the unit-wide level (on the <a href="#">Settings &gt; Content</a> page) and for any given conference <b>Content channel video</b> must also be enabled in the <a href="#">per-conference configuration</a>.</p>

Field	Field description	Usage tips
<b>View border size</b>	Select a border size for video transmitted to this endpoint.	<p>This sets a border thickness to display around the video image. This is useful where the image is displaying off the edges of the participant's screen; use a border to force the image to display properly.</p> <p>Applying a border size here means that this border size will always be used for this endpoint's transmitted video. Note that you can also apply a border to a participant in a conference from the <b>Conference list &gt; Conference &gt; Participant's display</b> page.</p>
<b>Default view family</b>	Sets the layout family to be used when calling out to this endpoint.	If this is set to <i>Use box-wide setting</i> then the default view family that has been configured via the <a href="#">Conference settings</a> page will be used.
<b>Preferred bandwidth from MCU</b>	The network capacity (measured in bits per second) used by the media channels established by the MCU to a single participant.	These settings take priority over the <b>Default bandwidth from MCU</b> setting configured in the global Conference settings (see <a href="#">Conference settings</a> ).
<b>Preferred bandwidth to MCU</b>	The maximum combined media bandwidth advertised by the MCU to endpoints.	These settings take priority over the <b>Default bandwidth to MCU</b> setting configured in the global Conference settings (see <a href="#">Conference settings</a> ).
<b>Layout control via FECC / DTMF</b>	<p>Whether this endpoint is able to change their view layout via Far-End Camera Control (FECC) or DTMF tones. Choose from:</p> <ul style="list-style-type: none"> <li>▶ <i>use conference configuration</i>: the setting for <b>Layout control via FECC / DTMF</b> will be determined by the configuration of the conference</li> <li>▶ <i>Disabled</i>: the participant using this endpoint will not be allowed to change their view layout using either FECC or DTMF</li> <li>▶ <i>FECC only</i>: the participant using this endpoint will be allowed to change their view layout using FECC</li> <li>▶ <i>FECC with DTMF fallback</i>: the participant using this endpoint will be allowed to change their view layout using FECC. If FECC is not available, this participant will be able to use DTMF</li> </ul>	This setting takes precedence over the per-conference layout control setting for conferences into which the endpoint is invited.
<b>Mute in-band DTMF</b>	<p>Use this option to mute in-band DTMF from this endpoint. Choose from:</p> <ul style="list-style-type: none"> <li>▶ <i>use conference configuration</i>: the setting for <b>Mute in-band DTMF</b> will be determined by the configuration of the</li> </ul>	In some scenarios, where a conference is cascaded onto an audio bridge, it might be useful for one of the participants in that conference to be able to send in-band DTMF to the MCU. This is for the purposes of sending the conference ID or PIN to the audio conferencing bridge. In

Field	Field description	Usage tips
	<p>conference</p> <ul style="list-style-type: none"> <li>▶ <i>Never</i>: the in-band DTMF from this endpoint will never be muted. Any DTMF tones sent from this endpoint will be audible to conference participants</li> <li>▶ <i>Always</i>: the in-band DTMF from this endpoint will always be muted. Any DTMF tones sent from this endpoint will not be audible to conference participants</li> <li>▶ <i>When used for MCU control</i>: if a participant is using in-band DTMF to control conference layout and for other in-conference features, the tones will be muted and will not be audible to conference participants. The MCU will only expect a participant to use DTMF for MCU control if <b>Layout control via FECC / DTMF</b> is set to <i>FECC with DTMF fallback</i> and <i>FECC is unavailable to the endpoint</i>.</li> </ul>	<p>this case, the <b>Mute in-band DTMF</b> setting for the endpoint of that participant needs to be <i>Never</i>. However, you can instead send DTMF tones to the audio conferencing bridge directly from the MCU; for more information refer to <a href="#">Sending DTMF to an audio bridge</a>.</p> <p>Unless you need to configure a particular setting for this endpoint, set this to <i>use conference configuration</i> and ensure you have the <b>Mute in-band DTMF</b> setting as required in the conference configuration (see <a href="#">Adding and updating conferences</a>).</p> <p>This setting takes precedence over the per-conference <b>Mute in-band DTMF</b> setting for conferences into which the endpoint is invited.</p>
<b>Initial audio status</b>	Set the initial audio status of an endpoint to be either active or muted.	If set to muted, when the endpoint joins a conference, it will not be able to contribute audio to the conference. For example, you can mute audio from an endpoint if somebody wants to be seen in the conference, but does not want to contribute verbally. You can mute both audio and video if required. This can be altered during the course of the conference by going to <b>Conferences</b> and clicking on the name of a conference and altering this participant's settings.
<b>Initial video status</b>	Set the initial video status of an endpoint to be either active or muted.	If set to muted, when the endpoint joins a conference, it will not be able to contribute video to the conference. For example, you can mute video from an endpoint if somebody wants to see the conference, but not be seen themselves. You can mute both audio and video if required. This can be altered during the course of the conference by going to <b>Conferences</b> and clicking on the name of a conference and altering this participant's settings.
<b>Automatic disconnection</b>	When a participant disconnects from a conference and only endpoints set to <b>Automatic disconnection</b> are left, all those participants are disconnected.	Set to enabled if you want this endpoint to be automatically disconnected from conferences when only endpoints set to <b>Automatic disconnection</b> remain in a conference when any other participant has disconnected.

Note that this setting is useful where you

Field	Field description	Usage tips
		<p>have configured a Cisco TelePresence IP VCR as an endpoint so that the IP VCR can be automatically called into a conference to record the session. In this case, the IP VCR will stop recording when the conference ends (that is, when everyone has left the conference). For more information about using the IP VCR in this way, refer to the IP VCR online help.</p>
<b>Custom codec selection</b>	Can be used to ensure only specific codecs are permitted on calls to (and received from) this endpoint.	<p>If <i>Enabled</i>, you can select which codecs are allowed to be used when communicating with this endpoint. When enabled this setting overrides the unit-wide codec selection on the <b>Settings &gt; Conference</b> page.</p>

## Configuring SIP endpoints

To configure the SIP endpoints to work with the MCU, go to **Endpoints > Add SIP**. This makes it easier to add endpoints to conferences because you can choose names from a list rather than adding network addresses.

Refer to the table below for tips on adding a SIP endpoint to the MCU. After entering the settings, click **Add endpoint**.

Field	Field description	Usage tips
<b>Name</b>	The name of the endpoint.	
<b>Address</b>	The IP address, host name, directory number, or SIP URI (in the format 1234@codian.com).	The address of the SIP endpoint can be a directory number if you are using a SIP registrar.
<b>Use SIP registrar</b>	Allows calls to this endpoint to use a directory number (in the <b>Address</b> field) and the SIP registrar.	If you have this enabled, you must configure the SIP registrar on the <b>Settings &gt; SIP</b> page.
<b>Outgoing transport</b>	Select the protocol to be used for call control messages for outgoing call connections to this endpoint.	<p>If you want this endpoint to use the unit-wide outgoing transport setting, select <i>box-wide default transport</i>. If this endpoint uses TCP, select TCP as the outgoing transport. If this endpoint uses UDP, select UDP as the outgoing transport. If this endpoint uses TLS, select TLS. Note that if you want the MCU to use TLS for call setup, you must have the encryption feature key (or the Secure management feature key) and the TLS service must be enabled on the <b>Network &gt; Services</b> page.</p> <p>Using TLS for call setup is not sufficient for the call to be considered encrypted such that it can participate in a conference which requires encryption. Where encryption is required in the conference configuration, a SIP call must use SRTP. For more information about SIP encryption, refer to <a href="#">Configuring encryption settings</a>.</p> <p>This setting overrides the unit-wide setting for <b>Outgoing transport</b> on the <b>Settings &gt; SIP</b> page. For more information about configuring SIP, refer to <a href="#">Configuring SIP settings</a>.</p>
<b>DTMF sequence</b>	The DTMF sequence to be sent to the audio conferencing bridge.	Allows the MCU to send DTMF tones to an audio bridge after the audio bridge has answered the call. In this way, the MCU can navigate audio menus. This is useful where a conference on the MCU dials out to an audio-only conference on an audio

Field	Field description	Usage tips
		<p>bridge.</p> <p>You can configure an audio bridge with a DTMF sequence as a pre-configured endpoint (either H.323 or SIP) which can then be added to any conference. Alternatively, you can add the audio bridge to an individual conference as a participant. You must specify the DTMF sequence in the Call-out parameters for that endpoint. The DTMF sequence can include digits 0-9 and * and #. There is a two second pause after the call connects after which the MCU will send the DTMF tones which are sent one every half second. You can insert as many additional two second pauses as you want by inserting commas into the DTMF sequence. Each comma represents a two second pause.</p> <p>For example, you want the MCU to dial out to a PIN-protected audio conference on an audio bridge. The conference ID is 555 and the PIN is 888. The audio bridge requires that you press # after entering the ID and after entering the PIN. In this example the DTMF sequence is: 555#,,888#. The two commas represent a four second pause which allows the audio bridge's automated menu system time to process the ID and request the PIN.</p>
<b>Call-in match parameters</b>	<p>These fields are used to identify incoming calls as being from the endpoint:</p> <ul style="list-style-type: none"> <li>• <b>Username:</b> This must be the username that the endpoint sends to the MCU</li> <li>• <b>IP address:</b> The IP address of the endpoint</li> </ul>	<p>The endpoint is recognized if all filled-in fields in this section are matched. Fields left blank are not considered in the match.</p> <p>Note that in some cases a SIP registrar can cause a call to appear to come from the IP address of the registrar rather than the IP address of the endpoint. In this case, to use call-in match parameters, leave the IP address field blank and enter the correct username. The call will be matched by username.</p> <p>When using LCS, the username that will be matched is the user's display name (e.g. Peter Rabbit) rather than the sign-in name (bluecoat@codian.com).</p>
<b>Display name override</b>	<p>The name that will be displayed in a conference as a label for this endpoint.</p>	<p>The name you enter here will override any default name configured on the endpoint. It will also override any other default name that might appear for an endpoint. For example, an endpoint's default name can be the name of the gateway through which the call was placed, or if the endpoint is</p>

Field	Field description	Usage tips
		<p>called-in via a gatekeeper, its E.164 number.</p> <p>Note that once an endpoint has connected, you cannot change the display name.</p>
<b>Motion / sharpness trade off</b>	<p>Choose whether to use the box-wide setting for motion/sharpness trade off, or configure an individual setting for this endpoint. Select from:</p> <ul style="list-style-type: none"> <li>▶ <i>Use box-wide setting:</i> this is the default value. In this case, the endpoint will use the motion/sharpness tradeoff setting from the <b>Settings &gt; Conferences</b> page</li> <li>▶ <i>Favor motion:</i> the MCU will try and use a high frame rate. That is, the MCU will strongly favor a resolution of at least 25 frames per second</li> <li>▶ <i>Favor sharpness:</i> the MCU will use the highest resolution that is appropriate for what is being viewed</li> <li>▶ <i>Balanced:</i> the MCU will select settings that balance resolution and frame rate (where the frame rate will not be less than 12 frames per second)</li> </ul>	<p>The settings for motion (frames per second) and sharpness (frame size or resolution) are negotiated between the endpoint and the MCU. This setting controls how the MCU will negotiate the settings to be used with this endpoint.</p>
<b>Transmitted video resolutions</b>	<p>Choose the setting for transmitted video resolutions from the MCU to this endpoint. This setting overrides the unit-wide setting on the <b>Settings &gt; Conferences</b> page.</p>	<p>Retain the default setting (<i>use box-wide setting</i>) unless you are experiencing problems with the display of certain resolutions by this endpoint.</p> <p>Endpoints advertise the resolutions that they are able to display. The MCU then chooses from those advertised resolutions, the resolution that it will use to transmit video. However, some endpoints do not display widescreen resolutions optimally. Therefore, you might want to use this setting to restrict the resolutions available to the MCU for transmissions to this endpoint.</p>
<b>Content contribution</b>	<p>Whether this endpoint is permitted to contribute the conference content channel. Select from:</p> <p>&lt;use conference default&gt;: this endpoint will use the <b>Content contribution from endpoints</b> setting from the per-conference configuration.</p> <p><i>Enabled:</i> This endpoint is allowed to contribute the content channel, even if content contribution from endpoints is</p>	<p>This setting is provided to allow you to individually configure whether or not an endpoint is allowed to contribute content to a conference.</p> <p>To use the content channel, the <b>Content status</b> must be enabled at the unit-wide level (on the <a href="#">Settings &gt; Content</a> page) and for any given conference <b>Content channel video</b> must also be enabled in the <a href="#">per-conference configuration</a>.</p>

Field	Field description	Usage tips
	<p>disabled in the per-conference configuration.</p> <p><i>Disabled:</i> This endpoint is not allowed to contribute the conference channel, even if content contribution from endpoints is enabled in the per-conference configuration.</p>	
<b>Content receive</b>		Currently, all content is sent to SIP endpoints in the main video channel.
<b>View border size</b>	Choose a border size for video transmitted to this endpoint.	<p>This sets a border thickness to display around the video image. This is useful where the image is displaying off the edges of the participant's screen; use a border to force the image to display properly.</p> <p>Applying a border size here means that this border size will always be used for this endpoint's transmitted video. Note that you can also apply a border to a participant in a conference from the <b>Conference list &gt; Conference &gt; Participant's display</b> page.</p>
<b>Default view family</b>	Sets the layout family to be used when calling out to this endpoint.	If this is set to <i>Use box-wide setting</i> then the default view family that has been configured via the <a href="#">Conference settings</a> page will be used.
<b>Preferred bandwidth from MCU</b>	The network capacity (measured in bits per second) used by the media channels established by the MCU to a single participant.	These settings take priority over the <b>Default bandwidth from MCU</b> setting configured in the global Conference settings (see <a href="#">Conference settings</a> ).
<b>Preferred bandwidth to MCU</b>	The maximum combined media bandwidth advertised by the MCU to endpoints.	These settings take priority over the <b>Default bandwidth to MCU</b> setting configured in the global Conference settings (see <a href="#">Conference settings</a> ).
<b>Layout control via FECC / DTMF</b>	<p>Whether this endpoint is able to change their view layout via Far-end Camera Control (FECC) or DTMF tones. Choose from:</p> <ul style="list-style-type: none"> <li>▶ <i>use conference configuration:</i> the setting for <b>Layout control via FECC / DTMF</b> will be determined by the configuration of the conference</li> <li>▶ <i>Disabled:</i> the participant using this endpoint will not be allowed to change their view layout using either FECC or DTMF</li> <li>▶ <i>FECC only:</i> the participant using this endpoint will be allowed to change their view layout using FECC</li> <li>▶ <i>FECC with DTMF fallback:</i> the participant using this endpoint will be</li> </ul>	This setting takes precedence over the per-conference layout control setting for conferences into which the endpoint is invited.

Field	Field description	Usage tips
	allowed to change their view layout using FECC. If FECC is not available, this participant will be able to use DTMF	
<b>Mute in-band DTMF</b>	<p>Use this option to mute in-band DTMF from this endpoint. Choose from:</p> <ul style="list-style-type: none"> <li>▶ <i>use conference configuration</i>: the setting for <b>Mute in-band DTMF</b> will be determined by the configuration of the conference</li> <li>▶ <i>Never</i>: the in-band DTMF from this endpoint will never be muted. Any DTMF tones sent from this endpoint will be audible to conference participants</li> <li>▶ <i>Always</i>: the in-band DTMF from this endpoint will always be muted. Any DTMF tones sent from this endpoint will not be audible to conference participants</li> <li>▶ <i>When used for MCU control</i>: if a participant is using in-band DTMF to control conference layout and for other in-conference features, the tones will be muted and will not be audible to conference participants. The MCU will only expect a participant to use DTMF for MCU control if <b>Layout control via FECC / DTMF</b> is set to <i>FECC with DTMF fallback and FECC is unavailable to the endpoint</i>.</li> </ul>	<p>In some scenarios, where a conference is cascaded onto an audio bridge, it might be useful for one of the participants in that conference to be able to send in-band DTMF to the MCU. This is for the purposes of sending the conference ID or PIN to the audio conferencing bridge. In this case, the <b>Mute in-band DTMF</b> setting for the endpoint of that participant needs to be <i>Never</i>. However, you can instead send DTMF tones to the audio conferencing bridge directly from the MCU; for more information refer to <a href="#">Sending DTMF to an audio bridge</a>.</p> <p>Unless you need to configure a particular setting for this endpoint, set this to <i>use conference configuration</i> and ensure you have the <b>Mute in-band DTMF</b> setting as required in the conference configuration (see <a href="#">Adding and updating conferences</a>).</p> <p>This setting takes precedence over the per-conference <b>Mute in-band DTMF</b> setting for conferences into which the endpoint is invited.</p>
<b>Initial audio status</b>	Set the initial audio status of an endpoint to be either active or muted.	If set to muted, when the endpoint joins a conference, it will not be able to contribute audio to the conference. For example, you can mute audio from an endpoint if somebody wants to be seen in the conference, but does not want to contribute verbally. You can mute both audio and video if required. This can be altered during the course of the conference by going to <b>Conferences</b> and clicking on the name of a conference and altering this participant's settings.
<b>Initial video status</b>	Set the initial video status of an endpoint to be either active or muted.	If set to muted, when the endpoint joins a conference, it will not be able to contribute video to the conference. For example, you can mute video from an endpoint if somebody wants to see the conference, but not be seen themselves. You can mute both audio and video if required. This can be altered during the course of the conference by going to <b>Conferences</b> and

Field	Field description	Usage tips
<b>Automatic disconnection</b>	When a participant disconnects from a conference and only endpoints set to <b>Automatic disconnection</b> are left, all those participants are disconnected.	clicking on the name of a conference and altering this participant's settings.  Set to enabled if you want this endpoint to be automatically disconnected from conferences when only endpoints set to <b>Automatic disconnection</b> remain in a conference when any other participant has disconnected.
<b>Custom codec selection</b>	Can be used to ensure only specific codecs are permitted on calls to (and received from) this endpoint.	If <i>Enabled</i> , you can choose which codecs are allowed to be used when communicating with this endpoint. When enabled, this setting overrides the unit-wide codec selection on the <b>Settings &gt; Conference</b> page.

## Configuring VNC endpoints

You can configure a VNC (Virtual Network Computing) endpoint on the MCU by choosing **Endpoints > Add VNC**. This makes it easier to add a VNC session to conferences because you can choose its name from a list rather than type its network address.

You can add VNC endpoints to the MCU. This enables you to have software applications running on a remote computer added to a conference. For example, if a speaker has a set of slides to present during a conference, you can add them as a VNC endpoint. After entering the settings, click **Add endpoint**.

Field	Field description	Usage tips
<b>Name</b>	The name of the computer used for the VNC application (VNC endpoint).	
<b>Host name or IP address</b>	The IP address or host name of the computer used for the VNC application (VNC endpoint).	
<b>Port</b>	The port used for communication to the VNC endpoint.	
<b>Password</b>	The VNC server password for the computer you're trying to view.	
<b>Re-enter password</b>	Re-enter the VNC password.	
<b>Automatic disconnection</b>	When a participant disconnects from a conference and only endpoints set to <b>Automatic disconnection</b> are left, all those participants are disconnected.	Set to enabled if you want this endpoint to be automatically disconnected from conferences when only endpoints set to <b>Automatic disconnection</b> remain in a conference when any other participant has disconnected.

## Displaying the gateway list

You can configure the Cisco TelePresence MCU to work with one or more H.323 gateways. The MCU can then effectively call through these configured gateways to one or more endpoints which are registered with the gateway but would not be reachable directly from the MCU.

For example, an IP PBX could be configured as a gateway, and the MCU could then call its registered E.164 numbers. An ISDN gateway can be configured as a gateway on the MCU allowing calls to ISDN endpoints and telephones.

You can add a Cisco TelePresence IP VCR as a gateway. This enables the MCU automatically call the IP VCR into a conference without the need for a gatekeeper; and in this way, the IP VCR can record directly into the folder you specify. For more information about using the IP VCR in this way, refer to the IP VCR online help.

The gateway list shows all of the currently configured H.323 gateways. To access this list, go to **Gateways**.

Field	Field description
<b>Name</b>	The descriptive name of the gateway.
<b>Address</b>	The IP address or host name of the gateway.
<b>Receive bandwidth</b>	The configured preferred bandwidth to the MCU from the gateway, or <i>&lt;default value&gt;</i> if no preference has been specified.
<b>Transmit bandwidth</b>	The configured preferred bandwidth from the MCU to the gateway, or <i>&lt;default value&gt;</i> if no preference has been specified.

To delete configured gateways, select the ones you want to delete and select **Delete selected**.

## Adding and updating gateways

You can configure the Cisco TelePresence MCU with one or more H.323 gateways:

- ▶ To add an H.323 gateway, go to **Gateways > Add new H.323 gateway**. After entering the settings described below, click **Add H.323 gateway**.
- ▶ To update an existing H.323 gateway, go to **Gateways** and click on a gateway name. After updating the settings described below, click **Update H.323 gateway**.

Field	Field description	More information
<b>Name</b>	The descriptive name of the gateway.	All gateways must have a unique name.
<b>Address</b>	The IP address or host name of the gateway.	
<b>Motion / sharpness trade off</b>	<p>Choose whether to use the unit-wide setting for motion/sharpness trade off, or configure an individual setting for this gateway. Choose from:</p> <ul style="list-style-type: none"> <li>▶ <i>Use unit-wide setting</i>: this is the default value. In this case, connections to the gateway will use the motion/sharpness tradeoff setting from the <b>Settings &gt; Connections</b> page</li> <li>▶ <i>Favor motion</i>: the MCU will try and use a high frame rate. That is, the MCU will strongly favor a resolution of at least 25 frames per second</li> <li>▶ <i>Favor sharpness</i>: the MCU will use the highest resolution that is appropriate for what is being viewed</li> <li>▶ <i>Balanced</i>: the MCU will select settings that balance resolution and frame rate (where the frame rate will not be less than 12 frames per second)</li> </ul>	The settings for motion (frames per second) and sharpness (frame size or resolution) are negotiated between the endpoint and the MCU. This setting controls how the MCU will negotiate the settings to be used with this endpoint.
<b>Preferred bandwidth from MCU</b>	The network capacity (measured in bits per second) used by the media channels established by the MCU to a single participant.	
<b>Preferred bandwidth to MCU</b>	Sets the bandwidth that the endpoint will advertise to the MCU when it calls it.	

## Displaying the built-in gatekeeper registration list

The MCU contains a built-in gatekeeper with which devices can register multiple IDs. IDs can be numbers, H.323 IDs (e.g. Fredsendpoint) or prefixes.

Up to 25 devices can be registered without a feature key. Feature keys can be purchased to increase this number.

Note: The MCU can register with its own built-in gatekeeper. The MCU then counts as one registered device. See [Configuring H.323 gatekeeper settings](#).

## Configuring the built-in gatekeeper

To start the gatekeeper:

1. Go to **Network > Services** and select **H.323 gatekeeper** to open a port for the gatekeeper. (On the MCU, ports are not open by default for security reasons.)
2. Go to **Gatekeeper**, select *Enabled* in the **Status** field and click **Apply changes**. If you attempt to enable the built-in gatekeeper without opening the port, an error message is displayed.

## Configuring neighboring gatekeepers

You can optionally configure the built-in gatekeeper with up to two neighboring gatekeepers. This means that if the built-in gatekeeper receives a request (known as an Admission Request or ARQ) to resolve an ID to an IP address and that ID is not currently registered with it then it will forward that request to its neighbor gatekeeper(s), as a Location Request (LRQ). The built-in gatekeeper will then use the information received from the neighbor(s) to reply to the original request.

You can also configure the behavior of the built-in gatekeeper on receipt of LRQs from another gatekeeper. It can:

- ▶ send LRQs regarding unknown IDs to its neighbor(s)
- ▶ reply to LRQs from other gatekeepers
- ▶ accept LCFs (Locations Confirms) from non-neighboring gatekeepers

Refer to this table for assistance when configuring the built-in gatekeeper:

Field	Field description	Usage tips
<b>Status</b>	Enables or disables the built-in gatekeeper.	To use the built-in gatekeeper, you must enable it here.
<b>Neighbor gatekeeper 1 and 2</b>	Enter the IP address(es), or hostname(s) (or <host>:<port number> to specify a port other than the default of 1719 on the neighboring gatekeeper), of the	These are the gatekeepers to which the built-in gatekeeper will send an LRQ if it has received an ARQ to resolve an ID which it does not currently have registered. The built-in gatekeeper will then use the

Field	Field description	Usage tips
	neighboring gatekeeper(s).	information received from the neighbor(s) to reply to the original request.
<b>Accept LRQs</b>	Configures the built-in gatekeeper to reply to LRQs from other gatekeepers.	These requests can come from any gatekeeper which has the MCU's built-in gatekeeper configured as one of its neighbors.
<b>Forward LRQs for unknown IDs</b>	<p>Configures the built-in gatekeeper to send (or not to send) LRQs regarding unknown IDs to its neighbor(s). Choose from the options:</p> <ul style="list-style-type: none"> <li>▶ <i>Disabled:</i> The MCU will only respond to LRQs about IDs registered with itself. It will not forward LRQs about IDs that are not registered with itself to neighboring gatekeepers.</li> <li>▶ <i>Enabled, using local return address:</i> The MCU will put, in the LRQ, its own address as the return address for the LCF.</li> <li>▶ <i>Enabled, using received return address:</i> The MCU will put, in the LRQ, the address of the gatekeeper that originated the request as the return address for the LCF. Use this option only if you are configuring the MCU to operate in an environment with a multiple-level gatekeeper hierarchy. For example, the 'received address' is required by the national gatekeepers connected to the Global Dialing Scheme (GDS).</li> </ul>	<p>Unless you have selected to <b>Accept LRQs</b>, you cannot configure the MCU to forward any LRQs.</p> <p>Enabling <i>using received return address</i> can be a significant security risk. Only use this setting with proper cause.</p>
<b>Accept LCFs from non-neighbors</b>	This setting enables the built-in gatekeeper to accept LCF message responses from any IP address.	<p>This setting is for use in environments with a multiple-level gatekeeper hierarchy. For example, this feature is required by the national gatekeepers connected to the Global Dialing Scheme (GDS).</p> <p>Enabling this setting can be a significant security risk. Only use this setting with proper cause.</p>

## Gatekeeper status

The number of registered devices is shown in the format X / Y where Y is the number of registered devices that your built-in gatekeeper is licensed for. Equally, the total number of registered IDs is

shown as Z / 1000, where 1000 is the maximum number of registrations allowed over all registered devices.

Below these summary figures is a table showing individual registrations. Registrations can be viewed by registered ID (the "ID view") or by device (the "Registration view"), giving complete and easily searchable lists. Switch between the views by clicking on the appropriate button.

The Registration view shows the summary per device (also known as the registrant), while the ID view shows individual registrations. This means that registrations from the same device are not necessarily listed together in the ID view but the view can be sorted by Registrant or Index to help you identify IDs belonging to the same registrant.

## ID view

Field	Field description	Usage tips
<b>ID</b>	The ID which the registrant has registered with the gatekeeper.	IDs can be numbers, H.323 IDs or prefixes.
<b>Type</b>	The type of registration.	One of: <i>E.164</i> (digits), <i>H.323 ID</i> or <i>Prefix</i> .
<b>Index</b>	This registrations index within the total number of registrations that this registrant has made with the gatekeeper.	In the format X / Y where Y is the number of registrations that this registrant has made with the built-in gatekeeper, and X is this particular registration's position within the total. Therefore, if a device registered 3 IDs with the gatekeeper and this was the second registration to be made, the Index would be 2 / 3.
<b>Registrant</b>	The IP address of the device from which this registration was made.	If the remote device has indicated via the RAI (Resource Availability Indication) mechanism that it is close to its resource limit, the Registrant will be labeled as "almost out of resources".

## Registration view

This view shows a one-line summary for each device registered with the built-in gatekeeper.

To deregister one or more devices (and all registrations for these devices), select the check boxes for the appropriate entries and then click **Deregister selected**.

Field	Field description	Usage tips
<b>Registrant</b>	The IP address of the device.	If the remote device has indicated via the RAI (Resource Availability Indication) mechanism that it is close to its resource limit, the Registrant will be labeled as "almost out of resources".
<b>H.323 ID</b>	The registered H.323 ID of the device.	To help identify registering devices, if the registrant has registered a H.323 ID (which will typically be its device name) that H.323 ID is shown here. If the device has registered multiple H.323 IDs, only the first is displayed.
<b>Registered IDs</b>	The number of registrations that this device has made with the built-in gatekeeper.	Click <b>(view)</b> to display individual registrations for the selected device. (The format is the same as the ID view, but the

Field	Field description	Usage tips
		table only includes entries for one device.)
<b>Registration time</b>	The time today or date and time of the last registration.	

## System defined users

The MCU is pre-configured with two user accounts ("admin" and "guest"), but you can also add other users (see [Adding and updating users](#)). Refer to the table below for descriptions of the pre-configured users.

User ID	Description	Usage tips
<b>admin</b>	<p>The MCU must have at least one configured user with administrator privileges. By default, the User ID is "admin" and no password is required.</p> <p>If you configure the MCU with advanced account security mode, a password is required. For more information about advanced account security mode, refer to <a href="#">Configuring security settings</a>.</p>	<p>After logging into the MCU for the first time (see <a href="#">Logging into the web interface</a>), you can change the User ID and password for this account. The privilege level is fixed at <i>administrator</i> for the admin user - who can see all the pages and change settings.</p>
<b>guest</b>	<p>The MCU must have at least one configured user with access privileges below <i>administrator</i>. The fixed User ID for this user is "guest" and by default no password is required.</p> <p>If you configure the MCU with advanced account security mode, the guest account requires a password that adheres to secure password criteria. For more information about advanced account security mode, refer to <a href="#">Configuring security settings</a>.</p>	<p>You cannot change the name of the "guest" User ID. You can add a password.</p>

You can modify the system defined user accounts if you need to. For example, for security, you should add a password to the admin account.

Note that you can also create new accounts with administrator or lower access privileges in addition to these pre-defined users (see [Adding and updating users](#)).

## User privilege levels

Every configured user in the MCU has an associated privilege level. There are seven defined privilege levels which determine the amount of control the user has over the MCU and its settings. Refer to the table below for details.

Privilege level	Access
<b>administrator</b>	<p>The main difference between an administrator and users with lower privilege levels is that administrators can change settings that affect all conferences and the configuration of the MCU itself, whereas other users only have access to individual conferences and to their own profiles.</p> <p>Users with administrator access can:</p> <ul style="list-style-type: none"> <li>▶ View MCU-wide status (<b>Status</b>)</li> <li>▶ Access all settings pages (<b>Settings</b>)</li> <li>▶ Perform software upgrades (<b>Settings &gt; Upgrade</b>)</li> <li>▶ Change system-wide conference settings (<b>Settings &gt; Conferences</b>)</li> <li>▶ View the Event log (<b>Logs</b>)</li> <li>▶ Configure H.323 gateways (<b>Gateways</b>)</li> <li>▶ Configure the built-in gatekeeper (<b>Gatekeeper</b>)</li> <li>▶ Manage users (<b>Users</b>)</li> <li>▶ Manage endpoints (<b>Endpoints</b>)</li> <li>▶ Configure auto attendants (<b>Conferences &gt; Auto attendants</b>)</li> <li>▶ Fully control conferences (<b>Conferences</b>)</li> </ul>
<b>conference creation and full control</b>	<p>Users with this privilege level can:</p> <ul style="list-style-type: none"> <li>▶ Change their own profile (<b>Profile</b>)</li> <li>▶ View the list of active conferences (<b>Conferences</b>)</li> <li>▶ View conferences via streaming (<b>Conferences</b>)</li> <li>▶ View participant lists for active conferences (<b>Conferences</b>)</li> <li>▶ Schedule new conferences (<b>Conferences</b>)</li> <li>▶ Fully control and modify all public conferences and the conferences they own (<b>Conferences</b>)</li> <li>▶ Configure H.323 gateways (<b>Gateways</b>)</li> <li>▶ Manage endpoints (<b>Endpoints</b>)</li> </ul>
<b>conference creation and limited control</b>	<p>Users with this privilege level can:</p> <ul style="list-style-type: none"> <li>▶ Change their own profile (<b>Profile</b>)</li> <li>▶ View the list of active conferences (<b>Conferences</b>)</li> <li>▶ View conferences via streaming (<b>Conferences</b>)</li> <li>▶ View participant lists for active conferences (<b>Conferences</b>)</li> </ul>

Privilege level	Access
	<ul style="list-style-type: none"> <li>▶ Schedule new conferences (<b>Conferences</b>)</li> <li>▶ Fully control and modify conferences they own (<b>Conferences</b>)</li> <li>▶ Exercise limited control of conferences owned by other users (<b>Conferences</b>)</li> </ul> <p>See <a href="#">Conference ownership</a> for additional information on which actions are permitted (and forbidden) by <b>limited control</b>.</p>
<b>conference creation</b>	<ul style="list-style-type: none"> <li>▶ Users with this privilege level can:</li> <li>▶ Change their own profile (<b>Profile</b>)</li> <li>▶ View the list of active conferences (<b>Conferences</b>)</li> <li>▶ View conferences via streaming (<b>Conferences</b>)</li> <li>▶ View participant lists for active conferences (<b>Conferences</b>)</li> <li>▶ Schedule new conferences (<b>Conferences</b>)</li> <li>▶ Fully control and modify conferences they own (<b>Conferences</b>)</li> </ul>
<b>conference detail</b>	<ul style="list-style-type: none"> <li>▶ Users with this privilege level can:</li> <li>▶ Change their own profile (<b>Profile</b>)</li> <li>▶ View the list of active conferences (<b>Conferences</b>)</li> <li>▶ View conferences via streaming (<b>Conferences</b>)</li> <li>▶ View participant lists for active conferences (<b>Conferences</b>)</li> </ul>
<b>conference list plus streaming</b>	<ul style="list-style-type: none"> <li>▶ Users with this privilege level can:</li> <li>▶ Change their own profile (<b>Profile</b>)</li> <li>▶ View the list of active conferences (<b>Conferences</b>)</li> <li>▶ View conferences via streaming (<b>Conferences</b>)</li> </ul>
<b>conference list only</b>	<ul style="list-style-type: none"> <li>▶ Users with this privilege level can:</li> <li>▶ Change their own profile (<b>Profile</b>)</li> <li>▶ View the list of active conferences (<b>Conferences</b>)</li> </ul>

## Displaying the user list

The **User list** page gives you a quick overview of all configured users on the MCU and provides a summary of some of their settings. To view the **User list** page, go to **Users**. Refer to the table below for assistance.

Field	Field description
<b>User ID</b>	The user name that the user needs to access the web interface of the MCU. Although you can enter text in whichever character set you require, note that some browsers and FTP clients do not support Unicode characters.
<b>Name</b>	The full name of the user.
<b>Privilege</b>	Access privileges associated with this user.  See <a href="#">User privileges</a> for detailed explanations.
<b>E.164</b>	The associated E.164 telephone number.
<b>Video endpoint</b>	The associated video endpoint.
<b>Picture</b>	The configured image to display for this user.

## Deleting users

To delete a user, select the user you want to delete and click **Delete selected users**. You cannot delete the admin and guest users.

# Adding and updating users

You can add users to and update users on the MCU. Although most information is identical for both tasks, some fields differ.

The MCU supports up to 200 users.

## Adding a user

To add a user:

1. Go to the **Users** page.
2. Click **Add user**.
3. Complete the fields referring to the table below to determine the most appropriate settings for the user.
4. After entering the settings, click **Add user**.

## Updating a user

To update an existing user:

1. Go to **Users**.

Field	Field description	More information
<b>User ID</b>	Identifies the log-in name that the user will use to access the MCU web interface.	Although you can enter text in whichever character set you require, note that some browsers and FTP clients do not support Unicode characters.
<b>Name</b>	The full name of the user.	
<b>Password</b>	The required password, if any.	<p>Although you can enter text in whichever character set you require, note that some browsers and FTP clients do not support Unicode characters.</p> <p>In advanced account security mode (configured on the <b>Settings &gt; Security</b> page), passwords must have:</p> <ul style="list-style-type: none"> <li>▶ at least fifteen characters</li> <li>▶ at least two uppercase alphabetic characters</li> <li>▶ at least two lowercase alphabetic characters</li> <li>▶ at least two numeric characters</li> <li>▶ at least two non-alphanumeric (special) characters</li> <li>▶ not more than two consecutive repeating characters. That is, two</li> </ul>

Field	Field description	More information
		<p>repeating characters are allowed, three are not</p> <p>In advanced account security mode, a password must be different from the previous ten used with that account. Also, a password will expire if it is not changed within 60 days.</p> <p>If the MCU is not using advanced account security mode, any password can be used.</p> <p>Note that passwords are stored in the configuration.xml file as plain text unless the MCU is configured (or has ever been configured) to use advanced account security mode. For more information, refer to <a href="#">Configuring security settings</a>.</p> <p>Note that this field is only active when adding a new user. If you are updating an existing user and want to change that user's password, click <b>Change password</b> instead.</p>
<b>Re-enter password</b>	Verifies the required password.	
<b>Disable user account</b>	Select to disable this account.	<p>This can be useful if you want to keep an account's details, but do not want anyone to be able to use it at the moment.</p> <p>You cannot disable the system-created admin account.</p> <p>The system-created guest account is disabled by default. If you enable it, the MCU will create a security warning.</p> <p>In advanced account security mode, a non-admin account will expire after 30 days of inactivity; that is, the MCU will disable it. To re-enable a disabled account, clear this option.</p> <p>For more information about advanced account security mode, refer to <a href="#">Configuring security settings</a>.</p>
<b>Lock password</b>	Prevents user from changing password.	This is useful where you want multiple users to be able to use the same user ID. The system-created guest account has <i>Lock password</i> enabled by default.
<b>Force user to change password on next login</b>	Select this option to force a user to change their password. Next time this user attempts to log in to the MCU, a change password prompt will appear.	This option is enabled by default for a newly created account. It is a good idea for new users to set their own secure

Field	Field description	More information
		<p>passwords.</p> <p>This option is not available for accounts where <i>Lock password</i> is selected.</p> <p>When the user changes his password, the MCU clears this check box automatically.</p>
<b>Privilege level</b>	The access privileges to be granted to this user.	See <a href="#">User privileges</a> for detailed explanations.
<b>E.164 phone number</b>	Associates an E.164 telephone number with a user account.	If the MCU receives a call from the E.164 phone number provided, it matches the number to the user account. This allows the MCU to take the appropriate action if one (or both) of the following fields are completed.
<b>Associated video endpoint</b>	Associates a configured endpoint with the user. This is used when a participant's video stream is from a separate device such as a web camera on a PC or a recording from a Cisco TelePresence IP VCR.	<p>If you set an associated video endpoint for a user, when a call is received from the E.164 phone number, the MCU knows that that call is audio-only. To provide the user with a video stream, the MCU calls the endpoint entered in this field. The user then has the complete conference experience with the audio on the telephone and a separate video stream for example on a computer with a web camera installed. The audio and video streams are matched so that the layout views for all participants reflect the level of audio received on the call from the E.164 phone number.</p> <p>Do not set the user's <b>E.164 phone number</b> and <b>Associated video endpoint</b> as the same number. If a user is going to be a normal video endpoint user, simply enter the E.164 phone number and leave the associated video endpoint field set to <i>None</i>.</p> <p>If required, a user's video contribution can be a recording on a Cisco TelePresence IP VCR. In this case the recording must first be configured as an H.323 endpoint on the MCU.</p> <p>See <a href="#">Adding an H.323 endpoint</a> for information about adding endpoints.</p>
<b>Picture upload</b>		
<b>Upload bitmap file</b>	The image to be used if a user joins a conference as an audio-only participant calling from the E.164 phone number specified above and there is no Associated video endpoint.	<p>This option is only available after you add the user.</p> <p>Because there is no video stream for the user, you can display a still image in the pane where the participant would normally</p>

Field	Field description	More information
		<p>appear.</p> <p>The maximum size of the picture is 176 x 144 pixels and the maximum file size is 500k.</p> <p>Click <b>Browse</b> to locate the image (jpg, gif, or Windows bmp file). Then, click <b>Send file</b> to upload the image to the MCU.</p> <p>The bitmap will only display if the user calls in to the MCU from the E.164 phone number using an audio-only device and there is no associated video endpoint.</p>

## Updating your user profile

You can make some changes to your user profile. To do this, go to **User profile**. Refer to the table below for tips.

Field	Field description	More information
<b>Name</b>	Your name, which identifies you to other users.	Changing this field does not change your log-in User ID.
<b>E.164 phone number</b>	Associates an E.164 telephone number with your user account.	<p>This limits the setup you will need to do each time you join a video conference. When the MCU receives a call from this number, it will be recognized as coming from your phone. If the device is an audio-only phone, you can set up an associated video endpoint and/or upload a picture file.</p> <p>This field is not available for the system <i>admin</i> account.</p>
<b>Associated video endpoint</b>	Associates a configured H.323 endpoint with your user account.	<p>If you call in to the MCU from your E.164 phone number using an audio-only device, the MCU calls your associated video endpoint and sends the conference video stream to that associated video endpoint (and receives a video-only stream from that endpoint).</p> <p>This field is not available for the system <i>admin</i> account.</p>
<b>Change password</b>		
<b>Current password</b>	Type your current password.	
<b>Password</b>	Type your new password.	<p>In advanced account security mode, passwords must have:</p> <ul style="list-style-type: none"> <li>▶ at least fifteen characters</li> <li>▶ at least two uppercase alphabetic characters</li> <li>▶ at least two lowercase alphabetic characters</li> <li>▶ at least two numeric characters</li> <li>▶ at least two non-alphanumeric (special) characters</li> <li>▶ not more than two consecutive repeating characters. That is, two repeating characters are allowed, three are not</li> </ul> <p>In advanced account security mode, a new password must be different to the previous 10 passwords that have been used with an</p>

Field	Field description	More information
		account.
<b>Re-enter password</b>	Verify your new password.	
<b>Picture upload</b>		
<b>Upload bitmap file</b>	You can upload an image which will display in the conference when you join conferences in audio-mode only.	<p>Click <b>Browse</b> to locate the bitmap image. Then, click <b>Send file</b> to upload the image to the MCU.</p> <p>The bitmap will only display if you call in to the MCU from your E.164 phone number using an audio-only device and you do not have an associated video endpoint.</p>

## Changing your password

In advanced account security mode, passwords must have:

- ▶ at least fifteen characters
- ▶ at least two uppercase alphabetic characters
- ▶ at least two lowercase alphabetic characters
- ▶ at least two numeric characters
- ▶ at least two non-alphanumeric (special) characters
- ▶ not more than two consecutive repeating characters. That is, two repeating characters are allowed, three are not

In advanced account security mode, a new password must be different to the previous 10 passwords that have been used with an account.

In advanced account security mode, if a user logs in with a correct but expired password the MCU asks that user to change the password. If the user chooses not to change it, that user is allowed two more login attempts to change the password before the account gets disabled.

In advanced account security mode, users other than administrator users are not allowed to change their password more than once in a 24 hour period.

If the MCU is not in advanced account security mode, there are no criteria for password selection.

If the MCU is in advanced account security mode, the above criteria for passwords are displayed on the **Change password** page.

# Configuring global conference settings

You can modify the global conference settings for the MCU choosing by **Settings > Conference**. However, many of these values can be overwritten by other MCU settings, for example individual conference, participant, or endpoint settings.

In this section:

- ▶ [Conference settings](#)
- ▶ [Advanced settings](#)

## Conference settings

Refer to this table for assistance configuring the conference settings. After making any configuration changes, click **Apply changes**.

Field	Field description	Usage tips
<b>Maximum video size</b>	Identifies the greatest video size that the MCU will send and receive when connected to a video endpoint.	This option is only available if the MCU 4CIF (HRO) feature key is present.
<b>Motion / sharpness trade off</b>	<p>Choose the unit-wide setting for motion/sharpness trade off. The options are:</p> <ul style="list-style-type: none"> <li>▶ <i>Favor motion</i>: the MCU will try and use a high frame rate. That is, the MCU will strongly favor a resolution of at least 25 frames per second</li> <li>▶ <i>Favor sharpness</i>: the MCU will use the highest resolution that is appropriate for what is being viewed</li> <li>▶ <i>Balanced</i>: the MCU will select settings that balance resolution and frame rate (where the frame rate will not be less than 12 frames per second)</li> </ul>	<p>The settings for motion (frames per second) and sharpness (frame size or resolution) are negotiated between the endpoint and the MCU. This setting controls how the MCU will negotiate the settings to be used with an endpoint.</p> <p>Note that the <b>Motion/sharpness trade off</b> setting for an individual endpoint will override this unit-wide setting for calls to/from that endpoint.</p>
<b>Transmitted video resolutions</b>	Choose the unit-wide setting for transmitted video resolutions. This setting can be overridden by individual configured endpoint settings.	<p>Retain the default setting (<i>Allow all resolutions</i>) unless you are experiencing problems with the display of certain resolutions by endpoints.</p> <p>Endpoints advertise the resolutions that they are able to display. The MCU then chooses from those advertised resolutions, the resolution that it will use to transmit video. However, some endpoints do not display widescreen resolutions optimally. In these cases, you might want to use this setting to restrict the resolutions available to the MCU.</p> <p>Note that you can configure this setting for individual configured endpoints if you do not need to restrict transmitted video</p>

Field	Field description	Usage tips
		resolutions for all endpoints.
<b>Default bandwidth from MCU</b>	Identifies the network capacity (measured in bits per second) used by the media channels established by the MCU to a single participant.	When the MCU makes a call to an endpoint, the MCU chooses the maximum bandwidth that is allowed to be used for the media channels which comprise that call. This field sets that maximum bandwidth, and is the total bandwidth of the audio, video, and content channels combined.  This setting can be overridden by individual endpoints' <b>Preferred bandwidth from MCU</b> values.
<b>Default bandwidth to MCU</b>	Sets the bandwidth that the MCU will advertise to the endpoint when it calls it.	This setting can be overridden by individual endpoints' <b>Preferred bandwidth to MCU</b> values.
<b>Default view family</b>	Determines which layout views (see <a href="#">Customizing layout views</a> ) new participants see when connecting to conferences.	Regardless of the family chosen here, participants can cycle through the available families using the Far End Camera Controls. See <a href="#">Understanding how participants display in layout views</a> .
<b>Use full screen view for two participants</b>	When there are only two participants, each participant will see the other in full-screen view.	If selected, when there are only two participants in a conference, this will apply regardless of which layout was originally chosen for the conference. If you do not select this setting, then the default family view is used with unused panes blank.
<b>Active speaker display</b>	When in a conference, there is generally one participant that the MCU recognizes as the <i>active speaker</i> , notionally the person currently speaking the loudest. This setting determines how the MCU displays that participant in conference views. <ul style="list-style-type: none"> <li>▶ <i>None</i> With this setting, no special action is taken when displaying the active speaker.</li> <li>▶ <i>Red border</i> Displays a red border around the active speaker.</li> <li>▶ <i>Green border</i> Displays a green border around the active speaker.</li> </ul>	Specifically, an active speaker is the participant who has been identified by the MCU as the current loudest speaker, and they are currently speaking. When you choose to have a border display around the active speaker, it will only display when that participant is speaking.
<b>Media port reservation</b>	Determines whether the MCU is operating in <i>Reserved mode</i> (Media port reservation <i>Enabled</i> ) or <i>Unreserved mode</i> (Media port reservation <i>Disabled</i> ).	See <a href="#">Port reservation modes</a> for additional information.
<b>Audio notifications</b>	Allows various audible in-conference features to be enabled or disabled.	<ul style="list-style-type: none"> <li>▶ The options are:</li> <li>▶ <i>Conference timing</i>: audible messages to indicate when the conference's scheduled end time is approaching.</li> </ul>

Field	Field description	Usage tips
		<ul style="list-style-type: none"> <li>▶ <i>Join and leave indications:</i> audible messages indicating when other participants join and leave the conference.</li> <li>▶ <i>Conference status:</i> audible status messages for example indicating to a participant that he is the only participant in a conference.</li> </ul>
<b>Overlaid icons</b>	Allows various in-conference icons to be displayed on participants' endpoints.	<ul style="list-style-type: none"> <li>▶ Depending on the check boxes that are selected, in-conference icons appear:</li> <li>▶ <i>Important participant:</i> a crown icon appears on all participants' endpoints in the pane of the participant that has become important.</li> <li>▶ <i>Unsecured conferences:</i> encrypted participants in a conference where encryption is optional see an icon indicating that there are other participants who are unencrypted: This icon is also visible to participants of an encrypted conference if there are people streaming that conference.</li> <li>▶ <i>Tunneled camera control:</i> an arrow icon appears on their endpoint when one participant uses the far end camera control to control another's camera.</li> <li>▶ <i>Layout changes:</i> an icon appears on their endpoint when a participant changes their layout view.</li> <li>▶ <i>Recording indicator:</i> an icon (a red dot) appears near the top left of the conference display to indicate that the conference is being recorded. For the recording indicator to display, the recording must be made by a Cisco TelePresence IP VCR running software version 2.1 or later, and the connection between the IP VCR and the MCU must be using H.323.</li> <li>▶ <i>Audio participants:</i> an icon appears near the top left of the conference display to indicate if there are any audio-only participants. To the right of the icon, the number of such participants will be displayed. If enabled, the icon will only display if there are one or more audio participants. Audio-only participants are participants that cannot be viewed; either the participant's endpoint cannot send</li> </ul>

Field	Field description	Usage tips
		<p>video, the MCU has not allocated a video port to the participant, or the participant has muted their endpoint's video channel to the MCU.</p> <ul style="list-style-type: none"> <li>▶ <i>Streaming participants</i>: an icon appears in the top left of the conference display to indicate if there are any streaming participants in the conference. If enabled, the icon will only display if there are one or more streaming participants. Streaming participants are people who are viewing and listening to the conference using a web browser.</li> <li>▶ <i>Media quality</i>: an icon (video camera with a cross through it) appears when a participant is experiencing high packet loss or if the network link's bandwidth is too low for the type of channel to the MCU that the endpoint has established.</li> <li>▶ Refer to <a href="#">Using in-conference features with video endpoints</a> to see all in-conference icons and their descriptions.</li> </ul>
<b>Overlaid text</b>	Allow various in-conference features to be enabled or disabled.	<ul style="list-style-type: none"> <li>▶ Depending on the check boxes that are selected, in-conference messages appear:</li> <li>▶ <i>Conference status</i>: messages appear when the conference's scheduled end time is approaching and when other participants join and leave the conference. Status messages can also be played for example when you are the only participant in a conference</li> <li>▶ <i>Conference timing</i>: a message appears when the conference's scheduled end time is approaching</li> <li>▶ <i>Join and leave indications</i>: messages appear when other participants join and leave the conference</li> <li>▶ <i>Text messages</i>: allows a message sent using the web interface to be displayed on participants' endpoints</li> <li>▶ <i>Content channel text chat</i>: messages that users can send one another via the content channel are displayed on participants' endpoints. Note that this functionality is only available if the web conferencing option (WCO) is activated on your MCU.</li> </ul>
<b>Conference</b>	Allows you to enter a message that will be seen by participants joining conferences on	The duration of the message is configured using the <b>Conference welcome message</b>

Field	Field description	Usage tips
<b>welcome message</b>	the MCU. The message is displayed at the bottom of a participant's conference display.	<b>duration</b> control.
<b>Conference welcome message duration</b>	This setting controls for how long (if at all) participants joining a conference will see the conference welcome message. Choose from: <ul style="list-style-type: none"> <li>▶ <i>&lt;never show&gt;</i></li> <li>▶ <i>5 seconds</i></li> <li>▶ <i>10 seconds</i></li> <li>▶ <i>1 minute</i></li> <li>▶ <i>&lt;permanent&gt;</i></li> </ul>	If you want streaming viewers to see the welcome message, you must configure the duration as <i>permanent</i> .
<b>Time to show participant names</b>	This setting controls whether (and for how long) participants shown in view panes are accompanied by their supplied name.	The "Conference welcome message" (described above) and any other overlaid textual messages (for instance information on how soon the conference is going to end, or endpoints leaving and joining the conference) will take priority over the displaying of participant names for the duration of those messages.

## Advanced settings

You typically only need to modify these advanced settings if you are working with a support engineer or setting up more complicated configurations.

Field	Field description	Usage tips
<b>Audio codecs from MCU</b>	Restricts the MCU's choice of audio codecs to be used for transmitting audio to endpoints.	When communicating with an endpoint, the MCU receives a list of supported audio codecs from the endpoint. The MCU chooses an audio codec from those available, and sends audio data to the endpoint in that format.  Note that the <b>Custom codec selection</b> setting for an individual endpoint will override this unit-wide setting for calls to that endpoint.
<b>Audio codecs to MCU</b>	Determines which audio codecs the MCU advertises to remote endpoints, restricting the endpoints' choice of channels available for sending audio data to the MCU.	Note that the <b>Custom codec selection</b> setting for an individual endpoint will override this unit-wide setting for calls from that endpoint.
<b>Video codecs from MCU</b>	Restricts the MCU's choice of video codecs to be used for transmitting video to endpoints.	When communicating with an endpoint, the MCU receives a list of supported video codecs from the endpoint. The MCU chooses a video codec from those available, and sends video data to the endpoint in that format.  Note that the <b>Custom codec selection</b>

Field	Field description	Usage tips
		setting for an individual endpoint will override this unit-wide setting for calls to that endpoint.
<b>Video codecs to MCU</b>	Determines which video codecs the MCU advertises to remote endpoints, restricting the endpoints' choice of channels available for sending video data to the MCU.	Note that the <b>Custom codec selection</b> setting for an individual endpoint will override this unit-wide setting for calls from that endpoint.
<b>Video transmit size optimization</b>	<p>Allows the MCU to vary the resolution and codec of the video being sent to a remote endpoint within the video channel established to that endpoint. The options are:</p> <ul style="list-style-type: none"> <li>▶ <i>None</i>: Do not allow video size to be changed during transmission</li> <li>▶ <i>Dynamic resolution only</i>: Allow video size to be optimized during transmission</li> <li>▶ <i>Dynamic codec and resolution</i>: Allow video size to be optimized during transmission and/or dynamic codec selection</li> </ul>	<p>With this option enabled, the MCU can, for instance, decide to send CIF video within a 4CIF channel if this will increase the viewed video quality.</p> <p>The circumstances under which decreasing the video resolution can improve the video quality include:</p> <ul style="list-style-type: none"> <li>▶ if the original size of the viewed video is smaller than the outgoing channel</li> <li>▶ if the remote endpoint has used flow control commands to reduce the bandwidth of the MCU video transmission</li> </ul> <p>Typically, lowering the resolution means that the MCU can transmit video at a higher frame-rate.</p>
<b>Video resolution selection mode</b>	<p>This setting can be used to influence the choice of outgoing video resolution made by the MCU in certain circumstances.</p> <ul style="list-style-type: none"> <li>▶ <i>Default</i> The MCU will use its normal internal algorithms to dynamically decide which resolution to send in order to maximize the received video quality.</li> <li>▶ <i>Favor 448p</i> The MCU will heavily favor sending 448p or w448p video (resolutions of 576 x 448 and 768 x 448 pixels respectively) to those endpoints that are known to work best with these resolutions.</li> </ul>	You should leave this at <i>Default</i> unless your environment dictates 448p or w448p resolutions only.
<b>Video format</b>	<p>Sets the format for video transmitted by the MCU.</p> <ul style="list-style-type: none"> <li>▶ <i>NTSC</i> The MCU will transmit video at 30 frames per second (or a fraction of 30, for example: 15fps)</li> <li>▶ <i>PAL</i> The MCU will transmit video at 25 frames per second (or a fraction of 25, for example: 12.5fps)</li> </ul>	<p>This option should be set to match your endpoints' video configuration. If you set this incorrectly, the smoothness of the video both to and from the endpoints might suffer.</p> <p>NTSC is typically used in North America, while PAL is typically used in the UK and Europe.</p>

Field	Field description	Usage tips
<b>Maximum transmitted video packet size</b>	Sets the maximum payload size (in bytes) of the packets sent by the MCU for outgoing video streams (from the MCU to connected video endpoints).	<p>Typically, you only need to set this value to lower than the default (1400 bytes) if there was a known packet size restriction in the path between the MCU and potential connected endpoints.</p> <p>Video streams generally contain packets of different lengths. This parameter only sets the <i>maximum</i> size of a transmitted network datagram. The MCU optimally splits the video stream into packets of this size or smaller. Thus, most transmitted packets will not reach this maximum size.</p>
<b>Interlaced video optimization</b>	Controls whether the MCU restricts video resolutions in order to reduce the effect of interlacing artifacts.	You should only enable this option if you are seeing video interlacing artifacts or on the advice of customer support. Note that all resolution restrictions imposed by this setting apply only to video being sent from endpoints to the MCU.
<b>Video receive bit rate optimization</b>	Enables the MCU to send bandwidth control messages to optimize the video bandwidth being used.	<p>The MCU can send these messages to endpoints requesting that the bandwidth of the video that they are sending be decreased or increased, up to the maximum bandwidth of the channel.</p> <p>If the participant is very prominent, then the MCU will ask the endpoint to send video at a high bandwidth. If the participant is not being viewed at all (or only being viewed in very small view panes), the MCU will request that the video is sent at a lower rate to conserve network bandwidth.</p>
<b>Flow control on video errors</b>	Enables the MCU to request that the endpoint send lower speed video if it fails to receive all the packets which comprise the far end's video stream.	<p>The MCU can send these messages to endpoints requesting that the bandwidth of the video that they are sending be decreased based on the quality of video received by the MCU.</p> <p>If there is a bandwidth limitation in the path between the endpoint and the MCU, it is better for the MCU to receive every packet of a lower rate stream than to miss some packets of a higher rate stream.</p>
<b>Don't see yourself in small panes</b>	Prevents the MCU from showing conference participants their own video in small panes of variable-sized pane views (and in conferences with equal-sized panes).	If this option is set, then a participant will never appear in a small pane (self-view), even if there is a free small pane available. They may still appear in larger panes, however, for example if the view focus is manually changed to show their video. See <a href="#">Understanding how participants display in layout views</a> for more details.
<b>Don't duplicate in small panes</b>	Prevents the MCU from duplicating large-pane participants in small panes.	When using a conference view with some large and some small panes, the MCU will typically duplicate in a small pane the

Field	Field description	Usage tips
		<p>video of a participant shown in a large pane. This is done to minimize the switching of small panes in response to changes of participant focus in the large pane. If you would prefer not to duplicate participants in small panes in this way, select this option. For more details of view layouts, see <a href="#">Understanding how participants display in layout views</a>.</p>
<p><b>Automatically make content channel important</b></p>	<p>Any new content channel in a conference will be treated as important and displayed prominently to all participants who see the content channel in their conference layout.</p>	<p>When this setting is enabled, any endpoint successfully contributing content to a conference is immediately treated as important. This has the same affect as using the 'crown' icon in the content channel row of a conference's <b>Participant list</b> page.</p> <p>An administrator can remove the importance from the content channel at any time in the conference.</p> <p>This setting does not affect participants who view the content channel independently from their conference panes (for example, those viewing the content channel on a separate video screen).</p> <p>This setting will not affect those participants using pane placement. Participants using pane placement who have not allocated a pane to the content channel, will not see the content channel even if it is 'important'.</p>
<p><b>Loudest speaker pane placement behavior</b></p>	<p>When pane placement is in use, this option affects the potential duplication of participants that are specifically placed in view panes with view panes configured to show the conference's current active speaker.</p> <ul style="list-style-type: none"> <li>▶ <i>Never duplicate placed participants</i> A pane set to show the loudest speaker will never show a participant that is specifically configured to be displayed in another layout pane. If another layout pane has been configured to show the participant which is the current active speaker, panes set to show the loudest speaker will instead show the conference's previous loudest speaker. If you never want a pane that is set to &lt;loudest speaker&gt; to duplicate a participant shown in another layout pane, choose this setting.</li> <li>▶ <i>Allow duplication of placed</i></li> </ul>	<p>For more details of view layouts, see <a href="#">Understanding how participants display in layout views</a>.</p>

Field	Field description	Usage tips
	<p><i>participants in small panes only</i> This is the default setting; panes configured to show the loudest speaker will be able to show participants that are configured to be displayed in one or more small panes for that layout, but not those shown in big panes. This is most appropriate when using layouts with more than one big pane, in order to make best use of the screen area.</p> <ul style="list-style-type: none"> <li>▶ <i>Allow duplication of placed participants in any pane</i> Panels set to &lt;loudest speaker&gt; will always show the current active speaker for a conference, whether or not any other layout panes have been specifically configured to show that participant.</li> </ul>	
<b>Pane rolling interval</b>	When pane placement is in use, this option determines how often panes set to "rolling" change which participant they are showing.	For more details of view layouts, see <a href="#">Understanding how participants display in layout views</a> .
<b>Voice switching sensitivity</b>	Determines how easy it is for a participant to replace the active speaker for a conference based on how loudly they are speaking.	A value of 0 means that it is very difficult for the active speaker to be replaced; a value of 100 means the active speaker can be replaced very easily.
<b>Additional audio delay relative to video</b>	Select the delay to be applied to audio relative to video.	A setting used only to rectify some types of 'lip sync' issues sometimes found with certain endpoints. Leave at default setting, unless advised to alter it by customer support or a reseller.
<b>Floor and chair control</b>	Controls whether connected endpoints are allowed to use floor and chair control operations.	<p>This option acts as a unit-wide control over whether any H.243 floor and chair control operations are permitted. Individual conferences may have their own floor and chair control policies set; however, if this setting is <i>Disabled</i> then no floor or chair control operations will be possible within conferences. Changing this setting will only affect new calls - existing calls will continue with the setting in force when they became active.</p> <p>For more information, refer to <a href="#">H.243 floor and chair control</a>.</p>
<b>Incoming calls to unknown conferences or auto attendants</b>	<p>Sets the default action when endpoints call into the MCU using an unknown E.164 number, conference number, or auto attendant. In other words, a number that does not correspond to any configured conference.</p> <ul style="list-style-type: none"> <li>▶ <i>Default auto attendant</i> The endpoint will enter the default</li> </ul>	This option can make it easier for callers to create ad hoc conferences if <i>Create new ad hoc conference</i> is selected. If you do not wish callers to be able to create conferences in this way, select one of the other options.

Field	Field description	Usage tips
	<p>auto attendant from which they may join existing conferences or potentially create a new conference (see <a href="#">Using an auto attendant</a>). This behavior is the same as if the endpoint had called the MCU using its IP address rather than a number.</p> <ul style="list-style-type: none"> <li>▶ <i>Disconnect caller</i> Endpoints are not allowed to call unknown conference or auto attendant numbers, and the call will be terminated.</li> <li>▶ <i>Create new ad hoc conference</i> A new conference will be created with the number called as its numeric identifier. The endpoint automatically joins this new conference. This option is not available if the MCU is in <a href="#">port reservation</a> mode.</li> </ul>	
<b>Failed preconfigured participants redial behavior</b>	Determines the redial options for failed preconfigured participants.	<p>Controls the frequency of redial attempts to preconfigured participants. The options are:</p> <ul style="list-style-type: none"> <li>▶ <i>Never redial</i> The MCU will never attempt to redial a preconfigured participant, even if they fail to connect.</li> <li>▶ <i>Default</i> The MCU will make nine retry attempts to a preconfigured participant who fails to connect. The first four will be at one minute intervals, the following five will be at five minute intervals.</li> <li>▶ <i>Constant redial</i> The MCU will make constant retry attempts to a preconfigured participant who fails to connect. The first four will be at one minute intervals, all redial attempts after this will be at five minutes.</li> </ul>
<b>Conferences remain locked when empty</b>	When enabled, conferences remain locked when all participants leave the conference.	Without this option selected, when the final participant leaves a locked conference, the MCU unlocks that conference.
<b>Use conference name as caller ID</b>	If enabled, when the MCU is calling out to an endpoint, the caller ID that the endpoint will see is the conference name.	Without this option selected, the caller ID is the name of the MCU. This setting applies to both H.323 and SIP endpoints.
<b>Require H.323 gatekeeper callers to enter PIN</b>	Instructs the MCU to request conference participants dialing into protected conferences using an E.164 number via an H.323 gatekeeper to enter a PIN before they may join the conference.	You may wish participants joining a conference via a gatekeeper not to need to enter a PIN, even for protected conferences. If this is the case, do not set this option. If you wish conferences to be

Field	Field description	Usage tips
		<p>protected, regardless of how participants connect, ensure you set this option.</p> <p>When this option is set, participants calling into a protected conference will be presented with PIN-entry screen instead of the normal conference view. The option has no effect for conferences with no PIN set.</p>
<b>Require a PIN for ad hoc conferences</b>	If this option is checked, a participant creating an ad hoc conference must enter a PIN for that conference. The MCU will not create the conference until the participant enters a PIN.	<p>When a PIN is required for ad hoc conferences, the auto attendant will wait forever for the participant to enter a PIN.</p> <p>This option is required in some highly secure environments.</p>
<b>Minimum required PIN length for ad hoc conferences</b>	The minimum number of digits required for a PIN.	This field is only available if <b>Require a PIN for ad hoc conferences</b> is selected. This forces a participant who is creating an ad hoc conference to protect it with a PIN with at least this number of digits.
<b>Time to wait when setting up ad hoc conference PIN</b>	<p>The timeout setting for a participant entering a PIN for an ad hoc conference that they are currently creating:</p> <ul style="list-style-type: none"> <li>▶ <i>&lt;never configure PIN&gt;</i> Participants will never be prompted to enter a PIN when creating an ad hoc conference.</li> <li>▶ <i>5 seconds, 10 seconds, 1 minute</i> Participants will be prompted for a PIN when creating an ad hoc conference. If the participant does not enter a PIN during the configured time period, the conference will be created without a PIN.</li> <li>▶ <i>&lt;wait forever&gt;</i> Participants will be prompted for a PIN when creating an ad hoc conference. They must either enter a PIN to create an ad hoc conference with a PIN or press the hash/pound key (#) to create the conference without a PIN.</li> </ul>	<p>This global setting may be useful where participants creating ad hoc conferences rarely need to configure a PIN.</p> <p>This field is unavailable when <b>Require a PIN for ad hoc conferences</b> is selected. When a PIN is required for ad hoc conferences, the auto attendant will wait forever for the participant to enter a PIN.</p>
<b>Advertise out of band DTMF to endpoints</b>	<p>If this option is checked, the MCU advertises the ability to receive out of band DTMF.</p> <p>If this option is checked, endpoints are allowed to send out of band DTMF. If this option is unselected, the MCU will not advertise the ability to accept out of band DTMF and endpoints will instead be forced to use in band DTMF.</p>	Prior to release 4.1, the MCU always advertised to endpoints the ability to receive out of band DTMF tones. Now you can disable this functionality if required. If you unselect this option, endpoints are forced to send DTMF in band (in the audio channel). This means that the MCU can pass DTMF tones on to an audio conferencing bridge or to another MCU where a conference is cascaded.

# Configuring H.323 gatekeeper settings

To configure gatekeeper settings, go to **Settings > H.323**.

You can configure the MCU to use a gatekeeper, which can make it easier for end-users to join conferences using directory numbers rather than requiring them to know the IP address or host name of the MCU. The MCU can register up to 100 IDs with the gatekeeper; these IDs comprise conferences' **Numeric IDs**, the **MCU service prefix**, and the **H.323 ID**. If you need to register more than 100 IDs, use a **prefix for MCU registrations** to route calls to the MCU, rather than registering individual conferences with the gatekeeper. The use of prefixes is described further in the table below.

In this section:

- ▶ [Gatekeeper settings](#)
- ▶ [Gatekeeper status](#)

## Gatekeeper settings

Refer to this table for assistance configuring the gatekeeper settings. After making any configuration changes, click **Apply changes**.

Field	Field description	Usage tips
<b>H.323 gatekeeper usage</b>	Enables the MCU to use an H.323 gatekeeper for registration of numeric identifiers for its conferences and/or auto attendants.	<p>When set to <i>Disabled</i> then no gatekeeper registrations are attempted (and existing registrations are torn down), regardless of other gatekeeper or per-conference settings.</p> <p>When set to <i>Enabled</i> registrations with the gatekeeper are attempted, and the gatekeeper is contacted for incoming and outgoing calls. If the gatekeeper does not respond, calls are still connected if possible. When set to <i>Required</i> registrations with the gatekeeper are attempted but calls are not connected if the gatekeeper cannot be contacted.</p>
<b>H.323 gatekeeper address</b>	Identifies the network address of the gatekeeper to which MCU registrations should be made.	<p>This can be specified either as a host name or as an IP address.</p> <p>This field will have no effect if <b>H.323 Gatekeeper usage</b> (see above) is set to <i>Disabled</i>.</p> <p>The gatekeeper can be either the built-in gatekeeper enabled on the <b>Gatekeeper</b> page (see <a href="#">Displaying the built-in gatekeeper registration list</a>) or an external gatekeeper. To use the built-in gatekeeper enter "127.0.0.1". For an external gatekeeper, enter its host name or IP address.</p>

Field	Field description	Usage tips
<b>Gatekeeper registration type</b>	Controls how the MCU identifies itself when registering with its configured gatekeeper.	<p>We recommend that you use the <i>Terminal / gateway</i> option unless you are using a service prefix (in this case, use <i>Gateway</i>). Only use a different option if you are:</p> <ul style="list-style-type: none"> <li>▶ having specific problems</li> <li>▶ using the Cisco TelePresence VCS Gatekeeper (with or without a service prefix), in which case use <i>MCU (standard)</i></li> <li>▶ using the Cisco Gatekeeper (with or without a service prefix), in which case use <i>Gateway (Cisco GK compatible)</i></li> <li>▶ using the VCON MXM Gatekeeper (with or without a service prefix), in which case use <i>MCU (compatible)</i></li> </ul> <p>Refer to the Knowledge Base in the Support section of the web site for more details about interoperability with gatekeepers.</p>
<b>Ethernet port association</b>	Whether a call involves consultation with the configured gatekeeper also depends on the <i>Port A</i> and <i>Port B</i> settings. For all incoming calls, and outgoing calls dialed by address rather than by E.164 phone number, the gatekeeper will be used to validate the connection only if the network port over which the connection is made is selected here.	
<b>(Mandatory) H.323 ID to register</b>	Specifies an identifier that the MCU can use to register itself with the H.323 gatekeeper.	<p>Before the MCU can register any IDs with the H.323 gatekeeper, it must make a unit-wide registration.</p> <p>This field is required for the gatekeeper registration.</p> <p>This will have no effect if <b>H.323 gatekeeper usage</b> is disabled.</p>
<b>Use password</b>	If the configured gatekeeper required password authentication from registrants, select the <b>Use password</b> check box and type the password.	Note that where password authentication is used, the <i>(Mandatory) H.323 ID to register</i> will be used as the username.
<b>Prefix for MCU registrations</b>	Specifies an optional group of digits that are added to the beginning of each conference or auto attendant's numeric identifier before registering it with the H.323 gatekeeper.	<p>Conferences and auto attendants registered with a gatekeeper have a <b>numeric identifier</b>. The numeric identifier is a unique sequence of digits entered from a video-conferencing endpoint to connect directly to the conference or auto attendant. This eliminates the need for users to navigate additional menus or to know the IP address of the MCU.</p> <p>To usefully partition the dialing space, you</p>

Field	Field description	Usage tips
		<p>might need to ensure that all registrations from a single MCU start with the same sequence of digits.</p> <p>Using registration prefixes also can benefit large-scale dial plan changes. For example, you can change all MCU registrations to begin with "121" instead of "11" by changing a single MCU configuration field rather than individually amending every conference or auto attendant's associated numeric identifier.</p> <p>If <b>H.323 gatekeeper usage</b> is disabled, this field will have no effect.</p> <p>Note that if you are also intending to use the MCU service prefix (see below), we recommend that you set both prefixes to the same number.</p>
<b>MCU service prefix</b>	If required, specify a group of digits which the H.323 gatekeeper may use to identify calls to be routed to the MCU.	<p>This field is optional. If set, users dialing any number beginning with this prefix will have their call directed to the MCU. This might be useful if you wish to create conferences in response to unknown E.164 numbers.</p> <p>Any numbers following the prefix will be identified by the MCU as a conference or auto attendant number. For example, if a conference has Numeric ID "3333" and you have set the service prefix to be "121", then a user dialing "1213333" will be connected to that conference.</p> <p>This field will have no effect if <b>H.323 gatekeeper usage</b> is disabled.</p> <p>Note that if you are also intending to use the <b>Prefix for MCU registrations</b> (see above), we recommend that you set both prefixes to the same number.</p> <p>If the MCU is unable to match a call to a conference or auto attendant, the action for <b>Incoming calls to unknown E.164 number</b> will be applied. This is a setting on the <b>Settings &gt; Conferences</b> page (see <a href="#">Configuring global conference settings</a>). This action can be set to <i>Create new ad hoc conference</i>.</p>
<b>Allow numeric ID registration for</b>	<p>This setting controls whether conferences' configured numeric IDs are allowed to be registered with the gatekeeper.</p> <p>This setting is the global control that <i>allows</i></p>	This would normally be <i>Enabled</i> ; by setting it to <i>Disabled</i> , an administrator can prevent users from adding to the set of IDs registered with the configured gatekeeper, even if those users enable the gatekeeper

Field	Field description	Usage tips
<b>conferences</b>	<p>or <i>disallows</i> conferences to be registered with the gatekeeper.</p> <p>Individual conferences will be registered or not depending on the per-conference <b>Numeric ID registration</b> setting which is on the <b>Add conference</b> page for scheduled conferences and for ad hoc conferences it is in the ad hoc conference template.</p>	<p><b>Numeric ID registration</b> setting for conferences they control. This may be desired when working with certain types of H.323 gatekeeper whose behavior in some modes is to disconnect active calls when the set of registered IDs changes.</p> <p>This setting affects both the registration of numeric IDs configured for scheduled conferences and the registration of ad hoc conferences.</p> <p>This field will have no effect if <b>H.323 Gatekeeper usage</b> is set to <i>Disabled</i>.</p>
<b>Send resource availability indications</b>	<p>Select this option if you want the MCU to inform the gatekeeper about its availability or non-availability. This information will be used by the gatekeeper when it is selecting where to place ad hoc conferences.</p> <p>Only use this option where multiple MCUs are registered with the <i>same</i> MCU service prefix on the same gatekeeper.</p> <p>If you select this option, you must configure the thresholds for conferences and/or video ports.</p> <p>Thresholds:</p> <ul style="list-style-type: none"> <li>▶ <b>Conferences:</b> Enter any number of conferences between 0 and 200. (A value of 0 will mean that the MCU will always indicate 'unavailable'.)</li> <li>▶ <b>Video ports:</b> Enter a number between 0 and the number of video ports on your MCU; for example, on an MCU 4205, there are 12 video ports, so enter a number between 0 and 12. (A value of 0 will mean that the MCU will always indicate 'unavailable'.)</li> </ul>	<p>The ability of the MCU to send resource availability messages is useful in a network where there are multiple MCUs or where there are several media blades in an MSE.</p> <p>In an environment with multiple conferencing devices registered with the same gatekeeper, that gatekeeper should favor devices in the available state when choosing where to place new calls.</p> <p>For example, when one MCU sends the gatekeeper a message indicating that it is not available, the gatekeeper will then attempt to use a different MCU for new ad hoc conferences.</p> <p>Resource availability indications are most useful where the thresholds are configured such that the MCU informs the gatekeeper that it is unavailable when its resources are nearly used up.</p> <p>Conferences without any active participants do not contribute to the conference count; any video port in use is added to the video port count.</p> <p>When either threshold is equaled or exceeded, the MCU sends a message to indicate that it is not available; when the resource usage drops such that neither threshold is equaled or exceeded, the MCU sends a message indicating that it is available.</p> <p>You might choose to only configure one of the thresholds. You are probably aware of how your video conferencing resources tend to be used by participants and you need to consider this when configuring the thresholds. For example, you could have</p>

Field	Field description	Usage tips
		four people who have each started an ad hoc conference; you might know that it is usual for such conferences to end up having ten participants. In this case, on a 40-port MCU you could set the conference threshold to 4 to indicate that it will be out-of-resources very soon. On the same MCU if you set the video port threshold to 35 and left the conference threshold empty, another four or more people could begin ad hoc conferences on this MCU before the participants expected in the original four conferences had dialed in.

## Gatekeeper status

The MCU also displays brief status information about its registrations with the configured gatekeeper.

To display a complete list of all IDs that the MCU is attempting to register with the configured H.323 gatekeeper, click the **details** link in the **Number of active registrations** row of the gatekeeper status table; this takes you to the [Active registrations](#) page.

Field	Field description	Usage tips
<b>H.323 gatekeeper status</b>	Displays the IP address of the gatekeeper currently being used by the MCU.	<p>This information might be useful if the gatekeeper has been specified with a host name rather than with an IP address.</p> <p>If the MCU has been unable to reach the configured gatekeeper and has instead registered with an alternate gatekeeper, the status displayed here will be "registered with alternate gatekeeper &lt;IP address&gt;".</p>
<b>Registered address</b>	Displays the local IP address and port number that the MCU has registered with the gatekeeper.	This information might be useful if the MCU has more than one IP address, for instance if both Ethernet interfaces are in use.
<b>Alternate gatekeepers available</b>	Displays the number of 'alternate' gatekeepers configured on the H.323 gatekeeper. This figure comes from the gatekeeper itself; if there are any 'alternate' gatekeepers configured, the gatekeeper tells the MCU their IP addresses.	<p>Where the configured gatekeeper has told the MCU about any configured 'alternate' gatekeepers and if the MCU loses contact with the configured gatekeeper, the MCU will attempt to register with each of the 'alternates' in turn. If none of the 'alternate' gatekeepers responds, the MCU will report that the registration has failed.</p> <p>If the MCU successfully registers with an 'alternate' gatekeeper:</p> <ul style="list-style-type: none"> <li>▶ the <b>H.323 gatekeeper status</b> will indicate that registration is with an 'alternate'</li> </ul>

Field	Field description	Usage tips
		<ul style="list-style-type: none"> <li>▶ the list of 'alternates' received from the new gatekeeper will replace the previous list</li> <li>▶ the MCU will only revert back to the original gatekeeper if the 'alternate' fails and only if the original gatekeeper is configured as an 'alternate' on the current gatekeeper's list of 'alternates'</li> </ul> <p>Note that if the MCU registers with an 'alternate' that does not itself supply a list of 'alternates', the MCU will retain the original list and if it loses contact with the current gatekeeper, each one will be attempted from the top again as before.</p>
<b>Resource availability status</b>	Displays whether the gatekeeper is configured to send resource availability indications and if it is, it displays the current state of the resource availability status of the MCU.	<p>The possible statuses are:</p> <ul style="list-style-type: none"> <li>▶ <i>resources available</i></li> <li>▶ <i>resources unavailable</i></li> <li>▶ <i>&lt;indications not configured&gt;</i></li> </ul>
<b>Number of active registrations</b>	Displays the number of E.164 numbers plus H.323 IDs plus prefixes that the MCU has registered with the gatekeeper.	<p>It also shows how many registrations are in progress but are not fully registered yet.</p> <p>Full information on the gatekeeper registrations being made by the MCU can be seen by clicking on <b>details</b>; this takes you to the <a href="#">Active registrations</a> page.</p>
<b>H.323 ID registration</b>	Displays the identifier that the MCU has used to register itself with the H.323 gatekeeper.	For more information about the H.323 ID, refer to the table above.
<b>MCU service prefix</b>	Displays the identifier that the gatekeeper has registered for calls to be routed to the MCU.	For more information about this prefix, refer to the table above.

# Displaying active gatekeeper registrations

To display a complete list of all IDs that the MCU is attempting to register with the configured H.323 gatekeeper, go to **Settings > H.323** and click **details**, shown next to the **Number of active registrations** status entry. You are taken to the **Active registrations** page. This page shows the complete set of IDs that the MCU is attempting to register with the configured H.323 gatekeeper, and includes the H.323 ID, prefixes, and specific E.164 number registrations for active conferences and configured auto attendants.

## Filters

You can configure filters so that only specific registrations are shown in the list. This may help you to find a registration whose number or name you know if the list is very long.

The filtered registration list is automatically updated when you change the ID and name filters; to stop filtering the list either delete the filters or click **Clear filters**. If both the ID filter and the name filter are defined, the registration list will show only those entries which match both filters.

Field	Field description	Usage tips
<b>ID filter</b>	Type the ID, or a part of the ID for which you want to see details.	The filtered registration list is automatically updated when you change the ID and name filters.
<b>Details filter</b>	Type the text, or a part of the text that will appear in the "Details" column of the Registrations table. For example, type 'Conference' to filter the registrations to show all conferences that the MCU is attempting to register with the gatekeeper.	

## Registration list

The registration list shows, for each registered ID, the type of that ID (H.323 ID, prefix or E.164 number), the object it relates to, and the status of that registration. If you want to modify or remove a specific registered ID, click on the link in its **Details** column to be taken to the relevant configuration page.

## Configuring SIP settings

To configure SIP settings on the MCU, go to **Settings > SIP**.

To allow conference participants with SIP endpoints to connect to the MCU by dialing a directory number rather than an IP address, you must configure a SIP registrar. The settings on this page control the MCU's interaction with the SIP registrar and with SIP endpoints. The MCU can register up to 50 URIs with the SIP Registrar; these URIs comprise conferences' Numeric IDs *and* the MCU itself. If you need to register more than 50 SIP URIs, use *prefix based call routing on SIP proxy* to route calls to the MCU, rather than registering individual conferences with the Registrar. Strip the prefix and leave only conference IDs in the URI before proxy the call to MCU.

A SIP call will select the audio and video codecs to use from the set of those both allowed on the **Settings > Conference** page and supported by SIP, unless the call is with an endpoint configured with a *Custom codec* (refer to [Configuring SIP endpoints](#) for more information).

Refer to this table for assistance configuring the SIP settings. After making any configuration changes, click **Apply changes**.

Field	Field description	Usage tips
<b>SIP registrar usage</b>	Enables the MCU to use a SIP registrar for registrations of numeric identifiers for its conferences and/or auto attendants.	<p>When set to <i>Disabled</i> then no SIP registrar registrations are attempted (and existing registrations are torn down), regardless of other per-conference settings.</p> <p>When set to <i>Enabled</i> registrations with the registrar are attempted, and calls can be made through the registrar. Note that 'Use SIP registrar' must be <i>Enabled</i> when adding a participant.</p>
<b>SIP registrar domain</b>	Identifies the network address of the SIP registrar to which MCU registrations should be made.	This can be specified either as a host name or as an IP address. This field will have no effect if <b>SIP registration settings</b> is set to <i>No registration</i> .
<b>SIP registrar type</b>	<p>Choose between:</p> <ul style="list-style-type: none"> <li>▶ <i>Standard SIP</i>: for non-Microsoft SIP registrars</li> <li>▶ <i>Microsoft OCS/LCS</i>: for Microsoft SIP registrars</li> </ul>	<p>Your choice is dependent on the type of SIP registrar you are using.</p> <p>This field will have no effect if <b>SIP registration settings</b> is set to <i>No registration</i>.</p>
<b>Username</b>	The login name for the MCU on the SIP registrar.	<p>You need to configure the SIP registrar with details of the devices that will register with it and create a login for each device.</p> <p>If you are using Microsoft OCS or LCS, you need to enter the full URI (for example, MCU@example.com).</p> <p>Note that this username will be used anywhere where one is required in a SIP call; for example, it will be used where authentication is required with a SIP server</p>

Field	Field description	Usage tips
		where no registrar is used.
<b>Password</b>	The password for the MCU on the SIP registrar.	<p>You need to configure the SIP registrar with details of the devices that will register with it and create a login for each device. The password configured on this page needs to match the password in the SIP registrar.</p> <p>Note that this username will be used anywhere where one is required in a SIP call; for example, it will be used where authentication is required with a SIP server where no registrar is used.</p>
<b>Allow numeric ID registration for conferences</b>	<p>This field controls whether any conferences' configured numeric IDs are allowed to be registered with the registrar.</p> <p>This setting is the global control that <i>allows</i> or <i>disallows</i> conferences to be registered with the SIP registrar.</p> <p>Individual conferences will be registered or not depending on the per-conference <b>Numeric ID registration</b> setting which is on the <b>Add conference</b> page for scheduled conferences and for ad hoc conferences it is in the ad hoc conference template.</p>	<p>This would normally be <i>Enabled</i>; by setting it to <i>Disabled</i>, an administrator can prevent users from adding to the set of IDs registered with the configured registrar, even if those users enable the registrar Numeric ID registration setting for conferences they control. This may be desired when working with certain types of SIP registrars whose behavior in some modes is to disconnect active calls when the set of registered IDs changes.</p> <p>This setting affects both the registration of numeric IDs configured for scheduled conferences and the registration of the numeric IDs of ad hoc conferences.</p> <p>This field will have no effect if SIP registrar usage is set to <i>Disabled</i>.</p>

## SIP call settings

Field	Field description	Usage tips
<b>SIP proxy address</b>	Identifies the network address of the SIP proxy.	If set, the proxy is used for all SIP calls, whether through a registrar or not.
<b>Maximum bit rate from Microsoft OCS/LCS clients</b>	Select a maximum bit rate to use from Microsoft OCS/LCS clients.	<p>Microsoft OCS/LCS clients will try to use the maximum bit rate that the MCU advertises during the initial call setup. In most scenarios, you will not want OCS/LCS clients to use the <i>Default bandwidth from MCU</i> that is configured on the <b>Settings &gt; Conferences</b> page (<a href="#">Configuring global conference settings</a>). Use this setting to select an appropriate bit rate for Microsoft OCS/LCS clients.</p> <p>&lt;limit disabled&gt; will cause the MCU to advertise the <i>Default bandwidth from MCU</i>.</p>
<b>Outgoing transport</b>	Identifies the protocol to be used for call control messages for outgoing call connections.	If your SIP devices use TCP, select TCP as the outgoing transport. If your SIP devices use UDP, select UDP as the

Field	Field description	Usage tips
		<p>outgoing transport. If you want to use encrypted SIP, select TLS. Note that if you want to use TLS, you must have the encryption feature key (or the Secure management feature key) and the TLS service must be enabled on the <b>Network &gt; Services</b> page.</p> <p>Note that this unit-wide setting can be overridden on a per-configured endpoint basis if you require different settings for individual endpoints. For more information about configuring SIP endpoints, refer to <a href="#">Configuring SIP endpoints</a>.</p> <p>Using TLS for call setup is not sufficient for the call to be considered encrypted such that it can participate in a conference which requires encryption. Where encryption is required in the conference configuration, a SIP call must use SRTP. For more information about SIP encryption, refer to <a href="#">Configuring encryption settings</a>.</p> <p>The MCU can accept connections on TCP, UDP, and TLS providing those services are enabled on the <b>Network &gt; Services</b> page (<a href="#">Configuring network services</a>).</p>
<b>Use local certificate for outgoing connections and registrations</b>	Select this option to force the MCU to present its local certificate when registering with the SIP registrar and when making outgoing TLS calls.	Often, the SIP registrar will not require the local certificate from the MCU. Only select this option if your environment dictates that the SIP registrar must receive the local certificate.

# Configuring streaming and ConferenceMe settings

You can choose to enable either streaming or the ConferenceMe software endpoint on your MCU.

To configure Streaming and ConferenceMe settings, go to **Settings > Streaming**.

To enable streaming, under **Streaming and ConferenceMe settings**, select *Streaming* in the **Enable** drop-down menu.

To enable ConferenceMe, under **Streaming and ConferenceMe settings**, select *ConferenceMe* in the **Enable** drop-down menu.

You can choose to have neither streaming nor ConferenceMe enabled on your MCU. To do this, select *None* from the **Enable** drop-down box.

Before your MCU is able to accept calls from ConferenceMe, a feature key must be installed on your MCU. ConferenceMe is included in the Web conferencing option. If you have purchased the Web conferencing option for your MCU, you do not need to install an additional feature key.

## Configuring streaming settings

You can configure two streaming options that will be available for users to choose between when streaming conferences. For each streaming option, you must select a media player and a bit rate.

You can configure the MCU to support either (or both of) unicast or multicast streaming. Unicast streaming involves a direct connection between the MCU and the individual user. Multicast streaming involves transmitting a single copy of the video or audio stream to multiple recipients. When choosing to support multicast streaming, ensure that your network has been properly configured to avoid network flooding.

Field	Field description	Usage tips
<b>Streaming &amp; ConferenceMe settings</b>		
<b>Enable</b>	Select whether to enable streaming, ConferenceMe, or neither.	<p>To enable streaming, select <i>Streaming</i> from the drop-down menu.</p> <p>To enable ConferenceMe, select <i>ConferenceMe</i> from the drop-down menu.</p> <p>If you would like to disable both streaming and ConferenceMe, select <i>None</i>.</p>
<b>Codec and bit rate choices</b>		
<b>Name</b>	The MCU is configured with two streaming rates, each comprising audio and video encoding formats (codecs) and overall (i.e. combined) media bit rate. The names you enter are the choices offered to users when	The name is optional - if not specified, the choices offered to users will simply show the details of the audio and video codecs, together with the overall media bit rate.

Field	Field description	Usage tips
	streaming.  This is intended to make the choice easier for potentially non-technical users; for instance, one rate could be named "low bit rate" and the other "high bit rate".	
<b>Player format</b>	Select from the available media players.	It is not possible to use Windows Media Player to stream conferences in multicast mode.
<b>Streams</b>	Sets the audio and video combination to use, and the overall media bit rate.	Typically, when two streaming rates are used, one is set up as a low bit rate combination and the other configured to use a higher bit rate. This enables those users with a low bandwidth connection to achieve useful streaming while at the same time allowing those with a high bandwidth link to take advantage of enhanced video.  The available options vary depending on which media player you have chosen.
<b>Multicast</b>	Choose whether to use the specified streaming rate for multicast.  If this is set to <i>Enabled</i> then for any conference with <b>Streaming</b> set to <i>Multicast</i> or <i>Unicast and multicast</i> , multicast audio and video streams will be transmitted according to the corresponding streaming settings.	Setting any <b>Multicast</b> parameters to <i>Enabled</i> will only be useful if the multicast media IP addresses and port numbers are also configured - see below.  It is not possible to use Windows Media Player to stream conferences in multicast mode.
<b>Multicast</b>		
<b>Media transmit IP address range</b>	Identifies the range of IP addresses to which multicast streaming media should be sent.	If this address range is not configured, the MCU cannot transmit multicast conference media.
<b>Media transmit port number range</b>	Sets the range of port numbers to which the media will be sent.	This UDP port number range is used in conjunction with the Media transmit IP address.  You must set both the start and end port numbers to transmit multicast conference media.
<b>Windows Media Player</b>		
<b>Protocol</b>	Identifies the protocol used by Windows Media Player to stream data. Note that v11 only supports HTTP; it no longer supports MMS over UDP or MMS over TCP.	MMS over UDP is a low latency/ high throughput protocol, while MMS over TCP includes extra reliability, though extra processing power is required. Both use the streaming port, so if a restrictive firewall is present HTTP may be necessary.  Auto negotiation attempts to open the most efficient connection, switching protocols if unable to do so, but on some systems this

Field	Field description	Usage tips
		is not efficient. We recommend that you select to use UDP/TCP/HTTP rather than auto-negotiation.

## Configuring ConferenceMe settings

Refer to this table for assistance setting up ConferenceMe. After you have completed the fields, click **Apply changes**.

Field	Field description	Usage tips
<b>Allow fall back to media using TCP</b>	Controls the ability of the MCU to connect using TCP if a UDP connection is not available.	ConferenceMe can connect using either UDP or TCP. When connecting using UDP, ConferenceMe will have better network performance, resulting in better video being sent to and received from the MCU.
<b>Maximum ConferenceMe connections</b>	The maximum number of ConferenceMe participants allowed on the MCU.	Allows you to define the number of ConferenceMe connections on your MCU, from a minimum of 0 to a maximum of 12. That is, you can have the defined number of participants in one conference or shared across several conferences.
<b>Maximum bandwidth from MCU</b>	The maximum bandwidth that can be sent from your MCU to ConferenceMe.	
<b>Maximum bandwidth to MCU</b>	The maximum bandwidth that can be received by your MCU from ConferenceMe.	

## Configuring content settings

The content settings affect the behavior of the MCU with regard to H.239 and BFCP (Binary Floor Control Protocol).

H.239 is the protocol that allows for an additional video channel (known as the content channel) alongside the main video channel in a video-conferencing call that uses H.323; BFCP is a protocol that allows for an additional video channel (known as the content channel) alongside the main video channel in a video-conferencing call that uses SIP. For example, a conference participant may want to contribute a slide presentation from a laptop within a video conference.

Note that BFCP uses the TCP port (5070); if you want to use BFCP, you must enable that service on the **Network > Services** page. For more information about configuring services, refer to [Configuring IP services](#).

For more information about content support in conferences, refer to [Content channel video support](#).

To access these settings, choose **Settings > Content**.

Refer to this table for assistance configuring the content settings. After making any configuration changes, click **Apply changes**.

Field	Field description	Usage tips
<b>Content status</b>	Controls whether the MCU as a whole is permitted to use content.	<p>If this setting is <i>Enabled</i>, you can still enable or disable the use of content on a per-conference basis. For more information about configuring individual conferences, refer to <a href="#">Adding and updating conferences</a>.</p> <p>If this setting is <i>Disabled</i>, no conference will be able to use content.</p> <p>If this setting is <i>H.239 only</i>, no conference will be able to use BFCP content.</p> <p>Certain video conferencing endpoints and infrastructure such as gatekeepers may not operate correctly when communicating with equipment (such as the Cisco TelePresence MCU) which declares H.239 capability. It may therefore be necessary to set this to <i>Disabled</i> in order to work with legacy devices (this will, of course, also prevent content video streams being used with any H.239- or BFCP-aware equipment).</p>
<b>Outgoing content video codec</b>	This field sets which video codec is used for conferences' content video streams.	A single content video stream is used for each conference, and this stream will be sent to all viewing endpoints. If an endpoint is unable to receive content video using the codec configured here, no content channel will be opened to that endpoint.

Field	Field description	Usage tips
<b>Minimum content channel bit rate</b>	<p>This field sets a lower limit on the bandwidth of the shared content video encoding sent to content receivers in a conference.</p> <p>Changing this setting when there are connected participants causes the MCU to re-assess whether there should be content video channels to those endpoints; the MCU will close existing channels and open new ones as appropriate.</p>	<p>A single content video stream is used for each conference, and this stream will be sent to all endpoints receiving the content channel as a separate video channel. If some endpoints are only able to receive low bit rate streams (for instance if they have called into the MCU at a low call rate), it is sometimes preferable to exclude those endpoints completely from the content stream rather than force all viewers to see a reduced bit rate channel.</p> <p>If you do not wish to exclude endpoints from viewing the shared content video channel in a conference, make sure this is set to <i>&lt;no minimum&gt;</i>, which is the default setting.</p> <p>Where an endpoint cannot, for whatever reason, receive the content channel as an additional video channel, the MCU can show the content channel as part of the main video channel. That is, the participant will see the content as a pane in the conference layout. This functionality is controlled by the <b>Display content in normal video channel</b> setting (see below).</p> <p>Note that during a call, an endpoint can send a 'flow control message' to the MCU that could cause the MCU to reduce the bit rate to that endpoint to below the configured <b>Minimum content channel bit rate</b>; in this case, the MCU will close the content channel to that participant. To re-enable content (which has been disabled in this way) to this participant, go to the conference's Participant list and use the content enable control. For more information about altering a participant's settings during a conference, refer to <a href="#">Viewing the conference participant list</a>.</p>
<b>Display content in normal video channel</b>	<p>Sets whether the MCU will render content channel data in endpoints' main video channels.</p>	<p>If there is an active content channel for a conference, it may be that the MCU is unable to open a content channel to a particular endpoint. For instance, that endpoint may have no content capability, or might not support the video format specified by the <b>Outgoing content video codec</b> setting (see above).</p> <p>In these cases, if this option is set to <i>Enabled</i>, the MCU will display the content channel video within a pane of the currently selected conference layout. In these cases, you might also want to enable the <b>Automatically make content channel</b></p>

Field	Field description	Usage tips
		<p><b>important</b> option in the <b>Settings &gt; Conferences</b> page which will make the content channel important at the same time.</p> <p>Note that SIP endpoints receive content from the MCU in the normal video channel (rather than in a separate channel). For this reason, this option must be enabled to allow SIP participants to see content.</p>
<b>Video fast update request filtering</b>	<p>If this option is <i>Enabled</i>, the MCU will ignore fast update requests for a conference's content video channel received from endpoints whose connections are experiencing problems.</p>	<p>When this mode is active, it can prevent a large number of keyframes being sent in the shared content video encoding in response to fast update requests from a single endpoint. A high number of keyframes may reduce the video quality of the content channel for all conference participants, including those with good connections to the MCU.</p>
<b>Web viewing applet bandwidth</b>	<p>Sets the bandwidth of the content channel video stream sent to streaming viewers.</p> <p>This option is only available if the MCU web conferencing (WCO) feature key is present.</p>	<p>Just as there is a single content video stream sent to all content-capable video conferencing endpoints in a conference, there is a single content channel video stream sent to all streaming viewers.</p> <p>This option allows the bandwidth of the streamed content channel video to be set. While in general a higher value means a greater frame rate (and thus better video quality), it is important to not set this value too high (i.e. higher than the available TCP network bandwidth between the MCU and viewers' machines) to avoid degradation resulting from lost data.</p>
<b>Markup of content channel video</b>	<p>If <i>Enabled</i>, people viewing the content channel stream via a web browser will be able to add markup to that video stream such as graphics and text. This markup will then be visible to all content channel viewers, both those viewing via streaming and those connected via videoconferencing endpoints.</p> <p>This option is only available if the MCU web conferencing (WCO) feature key is present.</p>	<p>Changing this setting has an immediate effect:</p> <ul style="list-style-type: none"> <li>▶ if changed to <i>Disabled</i>, it will not be possible for existing viewers to complete markup currently in progress</li> <li>▶ if changed to <i>Enabled</i>, existing streaming viewers will gain the ability to add markup</li> </ul>
<b>Automatic content handover</b>	<p>Allows for rapid swapping between content from different participants in a video conference. In a conference where a participant is sending content, this feature allows another endpoint to start sending content without having to wait for the current content provider to stop sending content from his computer.</p>	<p>The MCU applies this feature unit-wide. Therefore when it is enabled, all conferences on the MCU will allow automatic content handover.</p> <p>This option is disabled by default.</p>

# Configuring encryption settings

You can configure the MCU to encrypt connections to and from H.323 and SIP endpoints.

The encryption technology that the MCU uses for encryption to and from H.323 endpoints is Advanced Encryption Standard (AES).

The encryption technology that the MCU uses for encryption to and from SIP endpoints is Secure Real-time Transport Protocol (SRTP).

To use encryption, you must have the Encryption feature key present on the MCU. For information about installing feature keys, refer to [Upgrading the firmware](#). To access encryption settings, go to **Settings > Encryption**.

Encryption is used where both devices in a call agree to use encryption; by default if one of the devices cannot use encryption (for example if a SIP endpoint does not support SRTP), the MCU will allow the call to be unencrypted, unless the conference configuration dictates that encryption is *Required*. Where encryption is required, calls that cannot use encryption will not be allowed.

When encryption is in use to and from H.323 endpoints, the MCU will encrypt audio, video, and content media. It does not encrypt control or authentication information.

When encryption is in use to and from SIP endpoints, the MCU will encrypt audio and video media using SRTP. Control or authentication information can also be encrypted using TLS. For more information refer to [Using encryption with SIP](#), below.

You can:

- configure the MCU to advertise its ability to encrypt connections, such that it will use encryption if an H.323 endpoint can use AES encryption.
- configure the MCU to advertise its ability to encrypt connections, such that it will use encryption if a SIP endpoint can use SRTP encryption.
- configure the MCU so that the default encryption option for new conferences is either *Optional* or *Required*. Be aware that anyone creating a new conference will be able to set the encryption setting for the conference to either *Optional* or *Required*.
- force new ad hoc conferences to use encryption (by correctly configuring the ad hoc conference template, see [Using conference templates](#)).

Note that using encryption does not affect the number of ports that are available on the MCU.

Note that the MCU will not show thumbnail previews on the **Conference participant** page if encryption is required for a conference. If you have the **Show thumbnail images** option selected on the **Settings > User interface** page, thumbnail previews will be shown for conferences where encryption is optional and there are encrypted participants.

Refer to this table for assistance configuring the encryption settings. After making any configuration changes, click **Apply changes**.

Field	Field description	Usage tips
<b>Encryption status</b>	Whether the MCU is able to use encryption or not.	When encryption status is <i>Enabled</i> , the MCU advertises itself as being able to use encryption and will use encryption if required to do so by an endpoint. If this setting is <i>Enabled</i> , you can enable or

Field	Field description	Usage tips
		<p>disable the use of encryption on a per-conference basis.</p> <p>If this setting is <i>Disabled</i>, no conference will be able to use encryption.</p>
<b>SRTP encryption</b>	<p>Select the setting for media encryption for SIP calls:</p> <ul style="list-style-type: none"> <li>▶ <i>All transports</i>: If encryption is used for a call, the media will be encrypted using SRTP regardless of transport mechanism used for call control messages.</li> <li>▶ <i>Secure transports (TLS) only</i>: If encryption is used for a call, the media will only be encrypted in calls that are set up using TLS.</li> <li>▶ <i>Disabled</i>: SRTP will not be used for any calls. The MCU will not encrypt media for SIP calls.</li> </ul>	<p>For more information refer to <a href="#">Using encryption with SIP</a>, below.</p> <p>When disabled, the MCU will not advertise that it is able to encrypt using SRTP. It is only necessary to disable SRTP if it is causing problems.</p>

## Using encryption with SIP

The MCU supports the use of encryption with SIP. When encryption is in use with SIP, the audio and video media are encrypted using Secure Real-time Transport Protocol (SRTP). When using SRTP, the default mechanism for exchanging keys is Session Description Protocol Security Description (SDS). SDS exchanges keys in clear text, so it is a good idea to use SRTP in conjunction with a secure transport for call control messages. You can configure the MCU to also use Transport Layer Security (TLS) which is a secure transport mechanism that can be used for SIP call control messages.

Using TLS for call setup is not sufficient for the call to be considered encrypted such that it can participate in a conference which requires encryption. Where encryption is required in the conference configuration, a SIP call must use SRTP.

To configure the MCU to use SRTP to encrypt media in calls that are set up using TLS:

1. You must have the encryption feature key installed on your MCU.
2. Go to **Settings > Encryption** and set:
  - **Encryption status** to *Enabled*.
  - **Default setting for new scheduled conferences** to *Required*.
  - **SRTP encryption** to *Secure transports (TLS) only*.
3. Go to **Settings > SIP** and set **Outgoing transport** to *TLS*.

Note that to allow the MCU to accept incoming calls that use TLS, go to **Network > Services** and ensure that *Incoming Encrypted SIP (TLS)* is selected.

# Configuring security settings

To configure security settings, go to **Settings > Security**.

Field	Field description
<b>User authentication settings</b>	
<b>Advanced account security mode</b>	<p>Advanced account security mode causes the MCU to hash passwords before storing them in the configuration.xml file (see <a href="#">below</a>). Note that hashing user passwords is an irreversible process.</p> <p>Before you enable advanced account security mode, we recommend that you back up your configuration. The MCU gives you the option to do that after you have enabled Advanced account security mode.</p> <p>If you enable advanced account security mode, all current passwords (created when the MCU was not in advanced account security mode) will expire and users must change them.</p> <p>Advanced account security mode is described in greater detail <a href="#">below</a>.</p>
<b>Redirect HTTP requests to HTTPS</b>	<p>Enable this option to have HTTP requests to the MCU automatically redirected to HTTPS.</p> <p>This option is unavailable if either HTTP (<i>Web</i>) or HTTPS (<i>Secure web</i>) access is disabled on the <b>Network &gt; Services</b> page.</p>
<b>Idle web session timeout</b>	<p>The timeout setting for idle web sessions. The user must log in again if the web sessions expires. The timeout value must be between 1 and 60 minutes. Note that status web pages that auto-refresh will keep a web session active indefinitely. You can configure the MCU not to auto-refresh those pages; to do so, go to <b>Settings &gt; User interface</b> .</p>
<b>Serial console settings</b>	
<b>Hide log messages on console</b>	<p>The serial console interface displays log messages. If that is considered to be a security weakness in your environment, select this option to hide those messages.</p>
<b>Disable serial input during startup</b>	<p>Select this option for enhanced serial port security.</p>
<b>Require administrator login</b>	<p>Select this option to require an administrator login by anyone attempting to connect to the MCU via the console port. If this is not enabled, anyone with physical access to the MCU (or with access to your terminal server) can potentially enter commands on the serial console.</p>
<b>Idle console session timeout</b>	<p>If you have enabled <b>Require administrator login</b> , you can configure a session timeout period. The timeout setting for idle console sessions. The admin must log in again if the console sessions expires. The timeout value must be between 1 and 60 minutes.</p>

## Advanced account security mode

You can configure the MCU to use advanced account security mode. Advanced account security mode has the following features:

- ▶ The MCU will hash passwords before storing them in the configuration.xml file (see [below](#))
- ▶ The MCU will demand that passwords fulfill certain criteria, using a mixture of alphanumeric and non-alphanumeric (special) characters (see [below](#))
- ▶ Passwords will expire after 60 days
- ▶ A new password for an account must be different from the last ten passwords used with that account
- ▶ The MCU will disable a user's account if that user incorrectly enters a password three times consecutively. If this is an admin account, it is disabled for 30 minutes; for any other account, it is disabled indefinitely (or until you, the administrator, re-enable the account from the **User** page)
- ▶ Non-administrator account holders are not allowed to change their password more than once in any 24 hour period
- ▶ Administrators can change any user account's password and force any account to change its password by selecting **Force user to change password on next login** on the **User** page. Administrators can prevent any non-administrator account from changing its password by selecting **Lock password** on the **User** page.
- ▶ The MCU will disable any non-administrator account after a 30 day period of account inactivity. To re-enable the account, you must edit that account's settings on the **User** page

If you enable advanced security, all current passwords (created when the MCU was not in advanced account security mode) will expire and users must change them.

When using Advanced account security mode, we recommend that you rename the default administrator account. This is especially true where the MCU is connected to the public internet because security attacks will often use "admin" when attempting to access a device with a public IP address. Even on a secure network, if the default administrator account is "admin", it is not inconceivable that innocent attempts to log into the MCU will cause you to be locked out for 30 minutes.

We recommend that you create several accounts with administrator privileges. This will mean that you will have an account through which you can access the MCU even if one administrator account has been locked out.

If there are applications accessing the MCU, for example TMS, Conference Director, or any other API application, we recommend that you create dedicated administrator accounts for each application.

In advanced account security mode, if a user logs in with a correct but expired password the MCU asks that user to change the password. If the user chooses not to change it, that user is allowed two more login attempts to change the password before the account gets disabled.

## Hashing passwords

In advanced account security mode, the MCU will hash passwords before storing them in the configuration.xml file. The configuration.xml file is used for backing up and restoring the configuration of the MCU (see [Upgrading and backing up the MCU](#)). If you do not select to use advanced password security, all user passwords are stored in plain text in the configuration.xml; this might be a security issue. If you select to use advanced password security, they will not be stored anywhere on the MCU in plain text; instead the passwords will be stored as hash sums. Note that hashing user passwords is an irreversible process.

## Password format

In advanced account security mode, passwords must have:

- ▶ at least fifteen characters
- ▶ at least two uppercase alphabetic characters
- ▶ at least two lowercase alphabetic characters
- ▶ at least two numeric characters
- ▶ at least two non-alphanumeric (special) characters
- ▶ not more than two consecutive repeating characters. That is, two repeating characters are allowed, three are not

In advanced account security mode, a new password must be different to the previous 10 passwords that have been used with an account.

## Expiring passwords

In advanced account security mode, if a user logs in with a correct but expired password the MCU asks that user to change the password. If the user chooses not to change it, that user is allowed two more login attempts to change the password before the account gets disabled.

# Configuring network settings

To configure the network settings on the MCU and check the network status, go to **Network > Port A** or **Network > Port B**.

The MCU has two Ethernet interfaces, Port A and Port B. The configuration pages for the two interfaces look and behave similarly, and so are described together. Differences will be noted as appropriate.

Port A can be configured to be allocated its IP address by DHCP. Port B cannot use DHCP. Connect Port A to your local network and connect Port B to a second subnet or the internet depending on your application of the MCU.

In this section:

- ▶ [IP configuration settings](#)
- ▶ [IP status](#)
- ▶ [Ethernet configuration](#)
- ▶ [Ethernet status](#)

## IP configuration settings

These settings determine the IP configuration for the appropriate Ethernet port of the MCU. When you have finished, click **Update IP configuration** and then reboot the MCU.

Field	Field description	Usage tips
<b>IPv4 configuration</b>		
<b>IP configuration</b>	Specifies whether the port should be configured manually or automatically. If set to <i>Automatic via DHCP</i> the MCU obtains its own IP address for this port automatically via DHCP (Dynamic Host Configuration Protocol). If set to <i>Manual</i> the MCU will use the values that you specify in the Manual configuration fields below.	Click <b>Renew DHCP</b> to request a new IP address if you have selected automatic configuration.  Port A should never be disabled because it is the primary interface of the MCU.
<b>Manual configuration</b>		
<b>IP address</b>	The dot-separated IPv4 address for this port, for example 192.168.4.45.	You only need to specify this option if you have chosen <i>Manual</i> IP configuration, as described above.  For Port A, if the IP configuration setting is set to <i>Automatic by DHCP</i> this setting will be ignored.
<b>Subnet mask</b>	The subnet mask required for the IP address you wish to use, for example 255.255.255.0	
<b>Default</b>	The IP address of the default gateway on	

Field	Field description	Usage tips
<b>gateway</b>	this subnet, for example 192.168.4.1	
<b>DNS configuration</b>		
<b>Host name</b>	Specifies a name for the MCU.	Depending on your network configuration, you may be able to use this host name to communicate with the MCU, without needing to know its IP address.
<b>Name server</b>	The IP address of the name server.	
<b>Secondary name server</b>	Identifies an optional second name server.	The secondary DNS server is only used if the first is unavailable. If the first returns that it does not know an address, the secondary DNS server will not be queried.
<b>Domain name (DNS suffix)</b>	Specifies an optional suffix to add when performing DNS lookups.	This can allow you to use non-fully qualified host names when referring to a device by host name instead of IP address.  For example, if the domain name is set to <i>tandberg.com</i> , then a request to the name server to look up the IP address of host <i>endpoint</i> will actually lookup <i>endpoint.tandberg.com</i> .

## IP status

Use the IP status fields to verify the current IP settings for the appropriate Ethernet port of the MCU, which were obtained using DHCP or configured manually (see [IP configuration settings](#)) including:

- ▶ Host name
- ▶ DHCP
- ▶ IP address
- ▶ Subnet mask
- ▶ Default gateway
- ▶ Name server
- ▶ Secondary name server
- ▶ Domain name (DNS suffix)

## Ethernet configuration

These settings determine the Ethernet settings for the appropriate port of the MCU. Refer to the table for assistance with these settings. When you have finished, click **Update Ethernet configuration**.

Field	Field description	Usage tips
<b>Ethernet settings</b>	Specify whether you want this Ethernet port to automatically negotiate its Ethernet settings with the device it is connected to, or if it should use the values that you	It is important that your Ethernet settings match those of the device to which this port is connected. For example, both devices must be configured to use automatic

Field	Field description	Usage tips
	specify in the Manual configuration fields below.	negotiation, or both configured with fixed and matching speed and duplex settings (see below).
<b>Manual configuration</b>		
<b>Speed</b>	Identifies the connection speed: <i>10 Mbit/s</i> or <i>100 Mbit/s</i> . Use automatic negotiation if a connection speed of <i>1000 Mbit/s</i> is required.	The connection speed must match that of the device to which this port is connected.  You only need to select this option if you have chosen <i>Manual</i> Ethernet settings, as described above.
<b>Duplex</b>	Identifies the connection duplex mode: <ul style="list-style-type: none"> <li><i>Full duplex</i> Both devices can send data to each other at the same time</li> <li><i>Half duplex</i> Only one device can send to the other at a time</li> </ul>	The duplex setting must match that of the device to which this port is connected.  You only need to select this option if you have chosen <i>Manual</i> Ethernet settings, as described above.

## Ethernet status

Field	Field description	Usage tips
<b>Link status</b>	Indicates whether this Ethernet port is connected to or disconnected from the network.	
<b>Speed</b>	The speed ( <i>10/100/1000 Mbit/s</i> ) of the network connection to the MCU on this port.	This value is negotiated with the device to which this port is connected or based on your Manual configuration selected above.
<b>Duplex</b>	The duplex mode ( <i>Full duplex</i> or <i>Half duplex</i> ) of the network connection to this port.	This value is negotiated with the device to which this port is connected or based on your Manual configuration selected above.
<b>MAC address</b>	The fixed hardware MAC (Media Access Control) address of this port.	This value cannot be changed and is for information only.
<b>Packets sent</b>	Displays a count of the total number of packets sent from this port by the MCU. This includes all TCP and UDP traffic.	When troubleshooting connectivity issues, this information can help you confirm that the MCU is transmitting packets into the network.
<b>Packets received</b>	Displays a count of the total number of packets received by this port of the MCU. This includes all TCP and UDP traffic.	When troubleshooting connectivity issues, this information can help you confirm that the MCU is receiving packets from the network.
<b>Statistics:</b>	These fields display further statistics for this port. <ul style="list-style-type: none"> <li>▶ Multicast packets sent</li> <li>▶ Multicast packets received</li> <li>▶ Total bytes sent</li> <li>▶ Total bytes received</li> <li>▶ Receive queue drops</li> </ul>	Use these fields for advanced network diagnostics, such as resolution of problems with Ethernet link speed and duplex negotiation.

Field	Field description	Usage tips
	<ul style="list-style-type: none"><li data-bbox="359 264 528 293">▶ Collisions</li><li data-bbox="359 311 596 340">▶ Transmit errors</li><li data-bbox="359 358 587 387">▶ Receive errors</li></ul>	

# Configuring IP routes settings

If the *Video Firewall* feature is enabled (see [Upgrading the firmware](#)), you will need to set up one or more routing settings to control how IP traffic flows in and out of the MCU.

It is important that these settings are configured correctly, or you may be unable to make calls or access the web interface.

To configure the route settings, go to **Network > Routes**.

In this section:

- ▶ [Port preferences](#)
- ▶ [IP routes configuration](#)
- ▶ [Current IP status](#)

## Port preferences

If both Ethernet ports are enabled, it is necessary to specify which port is used in certain special circumstances. Make the appropriate selections described below. Click **Apply changes**.

Field	Field description	Usage tips
<b>Default gateway preference</b>	The IP address to which the MCU will send packets in the absence of more specific routing (see <a href="#">IP routes configuration</a> ). Therefore, it only makes sense to have precisely one default gateway, even though <i>different</i> default gateways may have been configured for Ports A and B. Use this option to decide which port's default gateway configuration to use as the unit's default gateway.	<p>If Ethernet Port B is disabled, you cannot specify that port as the default gateway preference.</p> <p>Selecting Port B as default gateway preference then disabling Port B will cause the preference to revert to Port A.</p>
<b>Name server (DNS) preference</b>	The IP address to which the MCU will send requests to look up unrecognized host names in order to determine their corresponding IP addresses. Only one name server (and associated secondary name server) may be used, even though <i>different</i> name servers may have been configured for Ports A and B. Use this option to decide which port's name server configuration to use as the unit's name server.	<p>If Ethernet Port B is disabled, you cannot specify that port as the name server preference.</p> <p>Selecting Port B as name server preference then disabling Port B will cause the preference to revert to Port A.</p>

## IP routes configuration

In this section you can control how IP packets should be directed out of the MCU. You should only change this configuration if you have a good understanding of the topology of the network(s) to which the MCU is connected.

Configuration of routes is divided into two sections: addition of new routes, and the display and removal of existing routes.

### Adding a new IP route

To add a new route, enter the details using the table below for reference. Click **Add IP route** to make the addition. If the route already exists, or aliases (overlaps) an existing route, you will be prompted to correct the problem and try again.

Field	Field description	Usage tips
<b>IP address / mask length</b>	<p>Use these fields to define the type of IP addresses to which this route applies.</p> <p>The IP address pattern must be in the dot-separated IPv4 format, while the mask length is chosen in the <b>IP address / mask length</b> field.</p> <p>The mask field specifies how many bits of the address are fixed; unfixed bits must be set to zero in the address specified.</p>	<p>To route all IP addresses in the range 192.168.4.128 to 192.168.4.255 for example, specify the IP address as 192.168.4.128 and the mask length as 25, to indicate that all but the last seven bits address are fixed.</p>
<b>Route</b>	<p>Use this field to control how packets destined for addresses matching the specified pattern are routed.</p>	<p>You may select <i>Port A</i>, <i>Port B</i> or <i>Gateway</i>. If <i>Gateway</i> is selected, specify the IP address of the gateway to which you want packets to be directed.</p> <p>Selecting <i>Port A</i> results in matching packets being routed to Port A's default gateway (see <a href="#">Configuring network settings</a>). Selecting <i>Port B</i> will cause matching packets to be routed to Port B's default gateway.</p> <p>If Ethernet Port B is disabled, the option to route packets to Port B will be disabled.</p>

### Viewing and deleting existing IP routes

Configured routes are listed below the **Add IP route** section. For each route, the following details are shown:

- ▶ The IP address pattern and mask
- ▶ Where matching packets will be routed, with the possibilities being:
  - ▶ Port A - meaning the default gateway configured for Port A
  - ▶ Port B - meaning the default gateway configured for Port B

- ▶ <IP address> - a specific address has been chosen
- ▶ Whether the route has been configured automatically as a consequence of other settings, or added by the user as described above.

The *default* route is configured automatically in correspondence with the *Default gateway preference* field (see [Port preferences](#)) and cannot be deleted. Any packets not covered by manually configured routes will be routed according to this route.

Manually configured routes may be deleted by selecting the appropriate check box and clicking **Delete selected**.

### Routes behavior with disabled ports

If the default gateway preference is set to Port B and that port is disabled, the default route will be updated automatically to route packets not covered by any manually configured route via Port A.

If a manually configured route specifies Port B and that port is disabled, packets matching that route **will not** be automatically routed via Port A, but discarded. You should take care to avoid this situation.

### Current IP status

This table shows the current default gateway and name server(s) for Ethernet Ports A and B. No fields can be changed, and are provided for reference when configuring the other parameters described in the sections above.

## Configuring IP services

To configure IP services, go to **Network > Services**.

Use this page to control the type of services that may be accessed via Ethernet Ports A and B. For example, if one Ethernet port is connected to a network outside your organization's firewall, and you wish to restrict the level of access that external users are entitled to, for example, by disabling FTP access via Port B. Refer to the table below for more details.

In addition to controlling the Ethernet interfaces over which a service operates, this page also allows an administrator to specify the port number on which that service is provided. If the port number for a service is changed, it is necessary to ensure that the new value chosen does not clash with the port number used by any of the other services; it is not, however, normally necessary to use anything other than the pre-configured default values.

Note that by default SNMP Traps are sent to port UDP port 162 (on the destination network management station); this is configurable. For more information, refer to [Configuring SNMP settings](#).

To reset all values back to their factory default settings, click **Reset to default** and then click **Apply changes**.

Field	Field description	Usage tips
<b>TCP service</b>		
<b>Web</b>	Enable/disable web access on the specified interface or change the port that is used for this service.	<p>Web access is required to view and change the MCU web pages and read online help files. If you disable web access on both Ports A and B you will need to use the serial console interface to re-enable it.</p> <p>Note that QuickTime uses RTSP by default which is listed as <b>Streaming (other)</b> on the <b>Network &gt; Services</b> page. However, the QuickTime player can be configured to use HTTP (that is it will come from the web service port) instead.</p> <p>If you require advanced security for the MCU, disable web access.</p> <p>If a port is disabled, this option will be unavailable.</p>
<b>Secure web</b>	Enable/disable secure (HTTPS) web access on the specified interface or change the port that is used for this service.	<p>This field is only visible if the MCU has the <i>Secure management (HTTPS)</i> feature key or an <i>Encryption</i> feature key installed. For more information about installing feature keys, refer to <a href="#">Upgrading and backing up the MCU</a>.</p> <p>By default, the MCU has its own SSL certificate and private key. However, you can upload a new private key and certificates if required. For more information about SSL certificates, refer to</p>

Field	Field description	Usage tips
		<p><a href="#">Configuring SSL certificates.</a></p> <p>If a port is disabled, this option will be unavailable.</p>
<b>Incoming H.323</b>	Enable/disable the ability to receive incoming calls to the MCU using H.323 or change the port that is used for this service.	<p>Disabling this option will not prevent outgoing calls to H.323 devices being made by the MCU.</p> <p>That is, the MCU will need to dial out to conference participants who are using H.323.</p> <p>If a port is disabled, this option will be unavailable.</p>
<b>Incoming SIP (TCP)</b>	Allow/reject incoming calls to the MCU using SIP over TCP or change the port that is used for this service.	<p>Disabling this option will not prevent outgoing calls to SIP devices being made by the MCU. That is, the MCU will need to dial out to conference participants who are using SIP over TCP.</p> <p>If a port is disabled, this option will be unavailable.</p>
<b>Incoming Encrypted SIP (TLS)</b>	Allow/reject incoming encrypted SIP calls to the MCU using SIP over TLS or change the port that is used for this service.	<p>Disabling this option will not prevent outgoing calls to SIP devices being made by the MCU. That is, the MCU will need to dial out to conference participants who are using SIP over TLS.</p> <p>If a port is disabled, this option will be unavailable.</p>
<b>BFCP</b>	Allow/reject content streams using BFCP (Binary Floor Control Protocol).	<p>Disabling this option will prevent BFCP content being used in any conferences. BFCP can be used to allow SIP endpoints to contribute and receive content. For more information, refer to <a href="#">Content channel video support.</a></p>
<b>Streaming (Windows Media Player)</b>	Allow/disable streaming from the MCU to Windows Media Player or change the port that is used for this service.	<p>If a port is disabled, this option will be unavailable.</p> <p>For more information about streaming, refer to <a href="#">Configuring streaming settings.</a></p>
<b>Streaming (other)</b>	Allow/disable RTSP (Real Time Streaming Protocol) streaming from the MCU to QuickTime or RealPlayer or change the port that is used for this service.	<p>If a port is disabled, this option will be unavailable.</p> <p>For more information about streaming, refer to <a href="#">Configuring streaming settings.</a></p>
<b>FTP</b>	Enable/disable FTP access on the specified interface or change the port that is used for this service.	<p>FTP can be used to upload and download MCU configuration.</p> <p>You should consider disabling FTP access on any port that is outside your organization's firewall.</p>

Field	Field description	Usage tips
		<p>If you require advanced security for the MCU, disable FTP access.</p> <p>If a port is disabled, this option will be unavailable.</p>
<b>UDP service</b>		
<b>SNMP</b>	<p>Enable/disable the receiving of the SNMP protocol on this port or change the port that is used for this service.</p>	<p>If a port is disabled, this option will be unavailable.</p> <p>You must use the same port number for both Port A and Port B. The number is automatically refreshed for Port B. You cannot change the Port B UDP port numbers and they are always grayed-out; if you want to enable the receiving of the SNMP protocol on Port B, ensure that you have the video firewall as an activated feature (refer to <a href="#">Upgrading and backing up the MCU</a>) and you have selected the check box for SNMP on Port B.</p> <p>Note that by default SNMP Traps are sent to port UDP port 162 (on the destination network management station); this is configurable. For more information, refer to <a href="#">Configuring SNMP settings</a>.</p> <p>If you require advanced security for the MCU, disable the SNMP service.</p>
<b>SIP (UDP)</b>	<p>Allow/reject incoming and outgoing calls to the MCU using SIP over UDP or change the port that is used for this service.</p>	<p>Disabling this option will prevent calls using SIP over UDP.</p> <p>If a port is disabled, this option will be unavailable.</p> <p>You must use the same port number for both Port A and Port B. The number is automatically refreshed for Port B. You cannot change the Port B UDP port numbers and they are always grayed-out; if you want to allow incoming and outgoing SIP (UDP) calls on Port B, ensure that you have the video firewall as an activated feature (refer to <a href="#">Upgrading and backing up the MCU</a>) and you have selected the check box for SIP (UDP) on Port B.</p>
<b>H.323 gatekeeper</b>	<p>Enable/disable access to the built-in H.323 gatekeeper or change the port that is used for the built-in H.323 gatekeeper.</p>	<p>If a port is disabled, this option will be unavailable.</p> <p>You must use the same port number for both Port A and Port B. The number is automatically refreshed for Port B. You cannot change the Port B UDP port numbers and they are always grayed-out; if you want to open Port B for the H.323</p>

Field	Field description	Usage tips
		gatekeeper, ensure that you have the video firewall as an activated feature (refer to <a href="#">Upgrading and backing up the MCU</a> ) and you have selected the check box for H.323 gatekeeper on Port B.
<b>Tunneled media</b>	Enable/disable tunneled media calls to the MCU over UDP	This option must be enabled to allow ConferenceMe to connect over UDP. This option is only available if you have the Web Conferencing feature key installed.

# Configuring SNMP settings

To configure monitoring using SNMP, go to **Network > SNMP**.

The MCU sends out an SNMP trap when the device is shut down or started up. The SNMP page allows you to set various parameters; when you are satisfied with the settings, click **Update SNMP settings**.

Note that:

- ▶ The 'system uptime' that appears in the trap is the time since SNMP was initialized on the MCU (and therefore will differ from the **Uptime** reported by the MCU on the **Status > General** page).
- ▶ The SNMP MIBs are read-only.

## System information

Field	Field description	Usage tips
<b>Name</b>	Identifies the MCU in the SNMP system MIB.	Usually you would give every device a unique name. The default setting is: Cisco Telepresence MCU
<b>Location</b>	The location that appears in the system MIB.	An optional field. It is useful where you have more than one MCU to identify where the unit is located. The default setting is: <i>Unknown</i>
<b>Contact</b>	The contact details that appear in the system MIB.	An optional field. The default setting is: <i>Unknown</i>  Add the administrator's email address or name to identify who to contact when there is a problem with the device. If SNMP is enabled for a port on the public network, take care with the details you provide here.
<b>Description</b>	A description that appears in the system MIB.	An optional field, by default this will indicate the model number of the unit. Can be used to provide more information on the MCU.

## Configured trap receivers

Field	Field description	Usage tips
<b>Enable traps</b>	Select this check box to enable the MCU to send traps.	If you do not select this check box, no traps will be sent.
<b>Enable authentication failure trap</b>	Select this check box to enable authentication failure traps.	You cannot select this check box unless you have selected to <b>Enable traps</b> above. Authentication failure traps are generated and sent to the trap receivers when someone tries to read or write a MIB value with an incorrect community string.
<b>Trap receiver addresses 1 to 4</b>	Enter the IP address or hostname for up to four devices that will receive both the general and the authentication failure traps.	The traps that are sent by the MCU are all SNMP v1 traps. You can configure trap receivers or you can view the MIB using a MIB browser. You can set the UDP port

Field	Field description	Usage tips
		number for the trap in the format <IP address>: <port number>. By default the UDP port number is 162.

## Access control

Field	Field description	Usage tips
<b>RO community</b>	Community string/password that gives read-only access to all trap information.	Note that SNMP community strings are not secure. They are sent in plain text across the network.
<b>RW community</b>	Community string/password that gives read/write access to all trap information.	It is advisable to change the community strings before enabling SNMP as the defaults are well known.
<b>Trap community</b>	Community string/password that is sent with all traps.	Some trap receivers can filter on trap community.

## Configuring QoS settings

To configure Quality of Service (QoS) on the MCU for audio and video, go to **Network > QoS**.

QoS is a term that refers to a network's ability to customize the treatment of specific classes of data. For example, QoS can be used to prioritize audio transmissions and video transmissions over HTTP traffic. These settings affect all audio and video packets to H.323 and SIP endpoints, and to streaming viewers. All other packets are sent with a QoS of 0.

The MCU allows you to set six bits that can be interpreted by networks as either Type of Service (ToS) or Differentiated Services (DiffServ).

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Note: Do not alter the QoS settings unless you need to do so.

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To configure the QoS settings you need to enter a six bit binary value.

Further information about QoS, including values for ToS and DiffServ, can be found in the following RFCs, available on the Internet Engineering Task Force web site [www.ietf.org](http://www.ietf.org):

- ▶ RFC 791
- ▶ RFC 2474
- ▶ RFC 2597
- ▶ RFC 3246

In this section:

- ▶ [About QoS configuration settings](#)
- ▶ [ToS configuration](#)
- ▶ [DiffServ configuration](#)
- ▶ [Default settings](#)

### About QoS configuration settings

The table below describes the settings on the **Network > QoS** page.

Click **Update QoS settings** after making any changes.

Field	Field description	Usage tips
<b>Audio</b>	Six bit binary field for prioritizing audio data packets on the network.	Do not alter this setting unless you need to.
<b>Video</b>	Six bit binary field for prioritizing video data packets on the network.	Do not alter this setting unless you need to.

## ToS configuration

ToS configuration represents a tradeoff between the abstract parameters of precedence, delay, throughput, and reliability.

ToS uses six out of a possible eight bits. The MCU allows you to set bits 0 to 5, and will place zeros for bits 6 and 7.

- ▶ Bits 0-2 set IP precedence (the priority of the packet).
- ▶ Bit 3 sets delay: 0 = normal delay, 1 = low delay.
- ▶ Bit 4 sets throughput: 0 = normal throughput, 1 = high throughput.
- ▶ Bit 5 sets reliability: 0 = normal reliability, 1 = high reliability.
- ▶ Bits 6-7 are reserved for future use and cannot be set using the MCU interface.

You need to create a balance by assigning priority to audio and video packets whilst not causing undue delay to other packets on the network. For example, do not set every value to 1.

## DiffServ configuration

DiffServ uses six out of a possible eight bits to set a codepoint. (There are 64 possible codepoints.) The MCU allows you to set bits 0 to 5, and will place zeros for bits 6 and 7. The codepoint is interpreted by DiffServ nodes to determine how the packet is treated.

## Default settings

The default settings for QoS are:

- ▶ *Audio 101110*:
  - For ToS, this means IP precedence is set to 5 giving relatively high priority. Delay is set to low, throughput is set to high, and reliability is set to normal.
  - For Diff Serv, this means expedited forwarding.
- ▶ *Video 100010*:
  - For ToS, this means IP precedence is set to 4 giving quite high priority (but not quite as high as the audio precedence). Delay is set to normal, throughput is set to high, and reliability is set to normal.
  - For DiffServ, this means assured forwarding (codepoint 41).

To return the settings to the default settings, click **Reset to default**.

# Displaying and resetting system time

The system date and time for the MCU can be set manually or using the Network Time Protocol (NTP).

To configure Time settings, go to **Settings > Time**.

Note that changing the time or NTP settings will have an effect on the recorded times in the Call Detail Records log. For more information, refer to [Working with Call Detail Records](#).

## System time

The current system date and time is displayed.

If you do not have NTP enabled and need to update the system date and/or time manually, type the new values and click **Change system time**.

## NTP

The MCU supports the NTP protocol. Configure the settings using the table below for help, and then click **Update NTP settings**.

The MCU re-synchronizes with the NTP server via NTP every hour.

If there is a firewall between the MCU and the NTP server, configure the firewall to allow NTP traffic to UDP port 123.

If the NTP server is local to Port A or Port B then the MCU will automatically use the appropriate port to communicate with the NTP server. If the NTP server is not local, the MCU will use the port that is configured as the default gateway to communicate with the NTP server, unless a specific IP route to the NTP server's network/IP address is specified. To configure the default gateway or an IP route, go to **Network > Routes**.

Field	Field description	Usage tips
<b>Enable NTP</b>	If selected, use of the NTP protocol is Enabled on the MCU.	
<b>UTC offset</b>	The offset of the time zone that you are in from Greenwich Mean Time.	You must update the offset manually when the clocks go backwards or forwards: the MCU does not adjust for daylight saving automatically.
<b>NTP host</b>	The IP address or hostname of the server that is acting as the time keeper for the network.	

## Using NTP over NAT (Network Address Translation)

If NAT is used between the MCU and the NTP server, with the MCU on the NAT's local network (and not the NTP server), no extra configuration is required.

If NAT is used between the MCU and the NTP server, with the NTP server on the NAT's local network, then configure the NAT forwarding table to forward all data to UDP port 123 to the NTP server.

# Upgrading and backing up the MCU

On this page:

- ▶ [Upgrading the main MCU software image](#)
- ▶ [Upgrading the loader software image](#)
- ▶ [Backing up and restoring the configuration](#)
- ▶ [Enabling MCU features](#)

## Upgrading the main MCU software image

The main MCU software image is the only firmware component that you will need to upgrade.

**To upgrade the main MCU software image:**

1. Go to **Settings > Upgrade**.
2. Check the **Current version** of the main software image to verify the currently installed version.
3. Log onto the [support pages](#) to identify whether a more recent image is available.
4. Download the latest available image and save it to a local hard drive.
5. Unzip the image file.
6. Log on to the MCU web browser interface.
7. Go to **Settings > Upgrade**.
8. Click **Browse** to locate the unzipped file on your hard drive.
9. Click **Upload software image**. The browser begins uploading the file to the MCU, and a new browser window opens to indicate the progress of the upload. When finished, the browser window refreshes and indicates that the "Main image upgrade completed."
10. The upgrade status displays in the **MCU software upgrade status** field.
11. [Shutting down and restarting the MCU](#).

## Upgrading the loader software image

Upgrades for the loader software image are not typically available as often as upgrades to the main software image.

**To upgrade the loader software image:**

1. Go to **Settings > Upgrade**.
2. Check the **Current version** of the loader software to verify the currently installed version.
3. Go to the software download pages of the web site to identify whether a more recent image is available.
4. Download the latest available image and save it to a local hard drive.
5. Unzip the image file.
6. Click **Browse** to locate the unzipped file on your hard drive.
7. Click **Upload software image**. The browser begins uploading the file to the MCU, and a new browser window opens to indicate the progress of the upload. When finished, the browser window refreshes and indicates that the "Loader image upgrade completed."
8. The upgrade status displays in the **Loader upgrade status** field.
9. [Shutting down and restarting the MCU](#).

## Backing up and restoring the configuration

The Back up and restore section of the **Upgrade (Settings > Upgrade)** page allows you to back up and restore the configuration of the MCU using the web interface. This enables you to either go back to a previous configuration after making changes or to effectively "clone" one unit as another by copying its configuration.

To back up the configuration, click **Save backup file** and save the resulting "configuration.xml" file to a secure location.

To restore configuration at a later date, locate a previously-saved "configuration.xml" file and click **Restore backup file**. When restoring a new configuration file to a MCU you can control which parts of the configuration are overwritten:

- ▶ If you select **Network settings**, the network configuration will be overwritten with the network settings in the supplied file. Typically, you would only select this check box if you were restoring from a file backed up from the same MCU or if you were intending to replace an out of service MCU. If you copy the network settings from a different, active, MCU and there is a clash (for instance, both are now configured to use the same fixed IP address) one or both boxes may become unreachable via IP. If you do not select **Network settings**, the restore operation will not overwrite the existing network settings, with the one exception of the QoS settings. QoS settings are overwritten regardless of the **Network settings** checkbox.
- ▶ If you select the **User settings** check box, the current user accounts and passwords will be overwritten with those in the supplied file. If you overwrite the user settings and there is no user account in the restored file corresponding to your current login, you will need to log in again after the file has been uploaded.

By default, the overwrite controls are not selected, and therefore the existing network settings and user accounts will be preserved.

Note that you can also back up and restore the configuration of the MCU using FTP. For more information, refer to [Backing up and restoring the configuration using FTP](#).

## Enabling MCU features

The MCU requires activation before most of its features can be used. (If the MCU has not been activated, the banner at the top of the web interface will show a prominent warning; in every other respect the web interface will look and behave normally.)

Advanced MCU features (such as *Video Firewall*) are not enabled as standard, and require additional activation. For information about configuring the video firewall, refer to the Knowledge Base section in the support pages of the web site.

If this is a new MCU you should receive the MCU already activated; if it is not, you have upgraded to a newer firmware version, or you are enabling a new feature, you may need to contact your supplier to obtain an appropriate activation code. Activation codes are unique to a particular MCU so ensure you know the unit's serial number such that you may receive a code appropriate to your MCU.

Regardless of whether you are activating the MCU or enabling an advanced feature, the process is the same.

### To activate the MCU or enable an advanced feature:

1. Check the **Activated features** (MCU activation is shown in this same list) to confirm that the feature you require is not already activated.

2. Enter the new feature code into the **Activation code** field exactly as you received it, including any dashes.
3. Click **Update features**. The browser window should refresh and list the newly activated feature, showing the activation code beside it. Activation codes may be time-limited. If this is the case, an expiry date will be displayed, or a warning that the feature has already expired. Expired activation codes remain listed, but the corresponding feature will not be activated. If the activation code is not valid, you will be prompted to re-enter it.
4. We recommend that you record the activation code in case you need to re-enter it in the future.

Successful MCU or feature activation has immediate effect and will persist even if the MCU is restarted.

Note that you can remove some MCU feature keys by clicking the **Remove** link next to the feature key in this page.

## Shutting down and restarting the MCU

It is sometimes necessary to shut down the MCU, generally to restart as part of an upgrade (see [Upgrading and backing up the MCU](#)). You should also shut down the MCU before intentionally removing power from it.

Shutting down the MCU will cause all conference participants to be disconnected, and allows the MCU to ensure that all data (such as Call Detail Records) is stored correctly. You will lose network connectivity with the MCU for a few minutes while you restart the unit.

### To shut down the MCU:

1. Go to **Settings > Shutdown**.
2. Click the **Shut down MCU** button.
3. Confirmation of shutdown is required; the button changes to **Confirm MCU shutdown**.
4. Click again to confirm.
5. The MCU will begin to shut down. The banner at the top of the page will change to indicate this.  
When the shutdown is complete, the button changes to **Restart MCU**.
6. Click this button a final time to restart the MCU.

## Displaying general status

The **General status** page displays an overview of the MCU status. To access this information, go to **Status > General**.

Refer to the table below for details of the information displayed

Field	Field Description
<b>System status</b>	
<b>Model</b>	The specific Cisco TelePresence MCU model.
<b>Serial number</b>	The unique serial number of the MCU.
<b>Software version</b>	The installed software version. You will need to provide this information when speaking to customer support.
<b>Build</b>	The build version of installed software. You will need to provide this information when speaking to customer support.
<b>Uptime</b>	The time since the last restart of the MCU.
<b>Host name</b>	The host name assigned to the MCU.
<b>IP address</b>	The IP address assigned to the MCU.
<b>CPU load</b>	The current processor utilization of the MCU.
<b>Media processing load</b>	<p>An overview of the current media loading of the MCU.</p> <p>If the total load is consistently high, you might need to add an additional MCU to better handle your video conferencing needs. Also, the total load may increase during periods of peak conference use. A low video load with high audio load implies that most of the conference participants have connected with audio-only endpoints.</p>
<b>System time</b>	
<b>Current time</b>	The system time on the MCU. Click <b>New time</b> to modify this value. The <b>Time Settings</b> page opens in which you can update the system date and time manually or refresh the time from an NTP server. For more information about the <b>Time Settings</b> page, refer to <a href="#">Displaying and resetting system time</a> .
<b>System log</b>	
<ul style="list-style-type: none"> <li>User requested shutdown</li> <li>User requested upgrade</li> <li>Unknown</li> </ul>	<p>The system log displays the last eight shutdown and upgrade events in date order with the most recent system log event at the top of the list.</p> <p>The log will also display "unknown" if there has been an unexpected reboot or power failure, which you should report to customer support if it happens repeatedly.</p>
<b>Diagnostic information</b>	
<b>Download diagnostic information</b>	If required to do so by customer support, click <b>Download diagnostic information</b> to save a set of diagnostic files.

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Field	Field Description
<b>Download conference information</b>	If required to do so by customer support, click <b>Download conference information</b> to save a file which details information about active and scheduled conferences for diagnostic purposes.

# Displaying conference status

The **Conference status** page displays the status of active and completed conferences and video and audio processing. To access this information, go to **Status > Conferences**.

Refer to the table below for assistance in interpreting the information displayed:

- ▶ [Conference status](#)
- ▶ [Video status](#)
- ▶ [Audio status](#)

For information about the number and type of ports provided by each MCU model, refer to [MCU port matrix](#).

## Format of displayed values

In many cases, the values displayed on this page are shown in the format **A (B) / C**; this represents:

- ▶ **A** – the current value of this statistic
- ▶ **B** – the maximum achieved value of this statistic (since last reset)
- ▶ **C** – the maximum allowable number for this statistic (this varies by MCU model)

Statistics for which there is no set maximum are displayed as **A (B)**, where **A** and **B** have the meanings described above.

Where the highest value attained is shown in parentheses (i.e. **B** in the above example), this value can be reset by selecting **Reset maximum values**. These values can be useful in monitoring peak MCU usage over a period of time.

## Conference status

Conference status displays an overview of active and completed conferences.

Field	Field description
<b>Active conferences</b>	The number of conferences that are currently configured on the MCU.
<b>Active auto attendants</b>	The number of auto attendants that are currently in use. If you dial in with an endpoint to the auto attendant, this will go up by one. It does not reflect the number of configured auto attendants.
<b>Completed conferences</b>	The number of conferences that were once active but are now not.
<b>Completed auto attendants</b>	The total number of calls into an auto attendant, excluding any in progress. If you call an auto attendant and enter into a conference or hang up the call, this number increases by one.
<b>Active conference participants</b>	The number of people currently in conferences.
<b>Previous conference participants</b>	The number of people who were previously participating in a conference (since the last time the MCU restarted).

Field	Field description
<b>Active streaming viewers</b>	The number of people currently watching conferences via a streaming application, such as Apple QuickTime or RealPlayer.
<b>TCP streaming viewers</b>	The number of streaming sessions out of the <b>Active streaming viewers</b> value shown above which are using TCP media transport rather than UDP.
<b>ConferenceMe users connected</b>	<p>The number of users currently connected to a conference using the ConferenceMe software endpoint. To change the number of allowed ConferenceMe participants, go to <b>Settings &gt; Streaming</b>.</p> <p>See <a href="#">Configuring ConferenceMe settings</a>.</p>
<b>Video ports in use</b>	This value is shown if the MCU is not operating in <a href="#">Port reservation mode</a> , and shows the number of video ports in use. This corresponds to the number of connected participants that are either contributing or being sent video, plus the number of conferences for which streaming is active.
<b>Audio-only ports in use</b>	This value is shown if the MCU is not operating in <a href="#">Port reservation mode</a> , and shows the number of audio-only ports in use. This corresponds to the number of connected participants that are contributing or being sent audio but not video.
<b>Streaming and content ports in use</b>	This value is shown if the model of MCU provides content channel and streaming ports (refer to <a href="#">MCU port matrix</a> ) and shows the total number of streaming and content ports in use across the currently active conferences.
<b>Reserved video ports</b>	<p>This value is shown if the MCU is operating in <a href="#">Port reservation mode</a>, and shows the total number of video ports reserved across the currently active conferences.</p> <p>Note that each conference for which streaming is enabled requires use of a video port or a streaming and content channel port (refer to <a href="#">MCU port matrix</a>). Where streaming and content channel ports are provided, streaming viewers and conferences' content channel video allocations will use the streaming and content channel ports rather than the available video ports; where streaming and content channel ports are not provided, streaming viewers and content channel allocations will use available video ports.</p> <p>If a streaming and content channel port or a video port is unavailable (or not allocated in advance when the MCU is in Reserved mode), it will not be possible to stream that conference. If a video port has been allocated for streaming a conference, any number of streaming viewers will be able to view that conference via streaming, at any combination of available bit rates. These streaming allocations are included in the total displayed.</p>
<b>Reserved audio-only ports</b>	This value is shown if the MCU is operating in <a href="#">Port reservation mode</a> , and shows the total number of audio-only ports reserved across the currently active conferences.
<b>Reserved streaming and content ports</b>	<p>This value is shown if the model of MCU provides content channel and streaming ports and which are operating in <a href="#">Port reservation mode</a>. For more information about the ports your MCU provides, refer to <a href="#">MCU port matrix</a>.</p> <p>This value shows the total number of streaming and content ports reserved across the currently active conferences.</p>

Field	Field description
<b>Reserved video ports in use</b>	This value is shown if the MCU is operating in <a href="#">Port reservation mode</a> , and shows, of the number of video ports reserved, how many are actually being used by active conference participants or streaming allocations.
<b>Reserved audio-only ports in use</b>	This value is shown if the MCU is operating in <a href="#">Port reservation mode</a> , and shows, of the number of audio-only ports reserved, how many are actually being used by active conference participants.
<b>Reserved streaming and content ports in use</b>	This value is shown if the model of MCU provides content channel and streaming ports and which are operating in <a href="#">Port reservation mode</a> . For more information about the ports your MCU provides, refer to <a href="#">MCU port matrix</a> .  This value shows the total number of streaming and content ports in use across the currently active conferences.

## Video status

Video status displays an overview of current video resource use.

Field	Field description	Usage tips
<b>Incoming video streams</b>	The number of video streams being received by the MCU.	Unicast indicates video streams sent directly to the MCU (incoming) or directly to the endpoints (outgoing) rather than multicast streams broadcast to the network and captured or sent by the MCU.
<b>Outgoing video streams</b>	The number of video streams being sent by the MCU.	
<b>Total incoming video bandwidth</b>	The total video data rate being received by the MCU.	
<b>Total outgoing video bandwidth</b>	The total video data rate being sent by the MCU.	

## Audio status

Audio status displays an overview of current audio resource use.

Field	Field description	Usage tips
<b>Incoming audio</b>	The number of audio streams being received by the MCU.	Unicast indicates audio streams sent directly to the MCU (incoming) or directly to

Field	Field description	Usage tips
<b>streams</b>		the endpoints (outgoing) rather than multicast streams broadcast to the network and captured or sent by the MCU.
<b>Outgoing audio streams</b>	The number of audio streams being sent by the MCU.	
<b>Complex (not G.711 or G.722) audio participants</b>	Displays active audio participants using neither G.711 nor G.722.	<p>On the MCU 4200 Series, the MSE 8420, and the MCU 4501 in SD mode, half of the MCU's allowable participants are permitted to use complex audio channels. On the MCU 4501 in HD or HD+ mode, all conference participants can use complex audio.</p> <p>A participant's endpoint is considered to be using complex audio if <i>either</i> it is transmitting a complex audio channel or the MCU is sending complex audio to it.</p>

# Conference content channel

The **Conference content channel** page shows various status items related to a conference's content channel. To view this page, go to **Conferences > Conference list**, click the name of the conference about which you want more information and on the **Participants** tab, click the Content channel link.

The displayed information is split into three sections:

## Preview

This section shows a graphical representation of the current content channel. If there is no active content channel, *inactive* will be shown here. Where there is an active content channel, clicking on this preview window will cause it to update.

## Received video

This section details the characteristics of the video stream supplying the content channel. This stream will either be a H.239 channel from one of the H.323 conference participants, or a BFCP (Binary Floor Control Protocol) channel from a SIP conference participant, or a *main video channel* configured for use as the content channel source (for instance, a VNC connection to a PC).

## Transmitted video

While there is at most one source video stream for a conference's content channel, the content channel can be viewed by several people, either via H.239 or BFCP to video conferencing endpoints, or via streaming (DJPEG) to desktop PCs. The **Transmitted video** section of this page shows the number of viewers of each type, plus some statistics on the currently active H.239/BFCP video stream and DJPEG channel.

## Diagnostics

It is possible to retrieve a set of diagnostics relating to the conference's content channel. This is accomplished by clicking on the **Download content channel diagnostics** control beneath the main table. You should not need to access these diagnostics except when directed to by customer support.

## Displaying hardware health status

The **Health status** page (**Status > Health**) displays information about the hardware components of the MCU.

Note: The **Worst status seen** conditions are those since the last time the MCU was restarted.

To reset these values, click **Clear**. Refer to the table below for assistance in interpreting the information displayed.

Field	Field description	Usage tips
<b>Fans</b> <b>Voltages</b> <b>RTC battery</b>	Displays two possible states: <ul style="list-style-type: none"> <li>▶ OK</li> <li>▶ Out of spec</li> </ul> States indicate both <i>Current status</i> and <i>Worst status seen</i> conditions.	The states indicate the following: <ul style="list-style-type: none"> <li>▶ <i>OK</i> – component is functioning properly</li> <li>▶ <i>Out of spec</i> – Check with your support provider; component might require service</li> </ul> If the <i>Worst status seen</i> column displays <i>Out of spec</i> , but <i>Current status</i> is <i>OK</i> , monitor the status regularly to verify that it was only a temporary condition.
<b>Temperature</b>	Displays three possible states: <ul style="list-style-type: none"> <li>▶ OK</li> <li>▶ Out of spec</li> <li>▶ Critical</li> </ul> States indicate both <i>Current status</i> and <i>Worst status seen</i> conditions.	The states indicate the following: <ul style="list-style-type: none"> <li>▶ <i>OK</i> – temperature of the MCU is within the appropriate range</li> <li>▶ <i>Out of spec</i> – Check the ambient temperature (should be less than 34 degrees Celsius) and verify that the air vents are not blocked</li> <li>▶ <i>Critical</i> – temperature of MCU is too high. An error also appears in the event log indicating that the system will shutdown in 60 seconds if the condition persists</li> </ul> If the <b>Worst status seen</b> column displays <i>Out of spec</i> , but <b>Current status</b> is <i>OK</i> monitor the status regularly to verify that it was only a temporary condition.

# Displaying security status

The Security status page displays a list of active security warnings for the MCU. To access this information, go to **Status > Security**.

Security warnings identify potential weaknesses in the security of the MCU's configuration. Note that some security warnings might not be relevant for your organization. For example if the MCU is inside a secure network, enabling HTTP may not be a security issue. For information about all possible security warnings, refer to [Understanding security warnings](#).

To acknowledge a security warning, select that warning and click **Acknowledge selected**. Acknowledged warnings will not appear on the MCU's Home page. If the MCU reboots, the warnings are reset and previously acknowledged warnings will need re-acknowledging.

To fix a security issue, click on the **Action** link for the warning message relating to the issue. When you fix a security issue, the security warning disappears from this list (on the **Status > Security** page), but it will be logged in the Audit log. For more information about the audit log, refer to [Working with the audit logs](#).

Refer to the table below for details of the information displayed.

Field	Field Description
<b>Warning</b>	The text of the security warning.
<b>State</b>	<p>For every security warning, the state will one of:</p> <ul style="list-style-type: none"> <li><i>New</i>: A new security warning is one that has been raised by the MCU, but you have not acknowledged it. New warnings also appear on the MCU Home page.</li> <li><i>Acknowledged</i>: An acknowledged security warning is one that you have acknowledged, but have not fixed.</li> </ul> <p>When you fix a security issue, the security warning disappears from this list, but it will be logged in the Audit log. For more information about the audit log, refer to <a href="#">Working with the audit logs</a>.</p>
<b>Action</b>	For every security warning, there is a corresponding action that explains how to fix the security issue. Usually this is a link that takes you to the page where you can make the configuration change that will fix the security issue.

## MCU port matrix

The port provision of the various MCU models is shown in the table below. Each video port can be used by one video-conferencing participant. Each audio-only port can be used by one voice-only participant in a video conference. Where streaming and content ports are provided, when streaming and content ports are used up, video ports will not be allocated to streaming viewers or providers of content; where streaming and content ports are not provided, conferences with streaming viewers and conferences with content will use available video ports.

For the MCU 4500 and the MSE 8510 Media2 blade, the number and type of available media ports on the MCU is controlled by the port capacity mode which you configure on the [Settings > Media ports](#) page.

- ▶ HD+ mode supports high definition video participants at up to 1080p at 30fps
- ▶ HD mode supports high definition video participants at up to 720p at 30fps
- ▶ SD mode supports participants at up to w448p at 30fps

### Port counts for clustered MSE Media2 blades

The maximum port counts for clusters comprising three MSE 8510 Media2 blades are as follows:

- ▶ For HD+ mode, the maximum number of port licenses that a three-blade cluster can utilize is 240. This will provide you with a total of 60 HD+ ports.
- ▶ For HD mode, the maximum number of port licenses that a three-blade cluster can utilize is 120. This will provide you with a total of 60 HD ports.

Note that in a cluster of MSE Media2 blades, SD ports are not available.

### Port matrix table

Model	Mode	Licenses per port	Video ports	Audio-only ports	Streaming and content ports
MCU 4203			6	6	6
MCU 4205			12	12	0
MCU 4210			20	20	0
MCU 4215			30	30	0
MCU 4220			40	40	0
MCU 4501	SD		12*	12*	6*
	HD		6*	6*	6*
	HD+		3†	6*	0
MCU 4505	HD		12	12	12
	HD+		6†	12	0
MCU 4510	HD		20	20	20
	HD+		10†	20	0
MCU 4515	HD		30	30	30

Model	Mode	Licenses per port	Video ports	Audio-only ports	Streaming and content ports
	HD+		15†	30	0
<b>MCU 4520</b>	HD		40	40	40
	HD+		20†	40	0
<b>MSE 8420 Media blade</b>		1	40	40	0
<b>MSE 8510 Media2 blade</b>	SD	1	80	0	0
	HD	2	20	20	20
	HD+	4	20	20	0

\* Can be doubled using the MCU 6 to 12 HD port upgrade feature key

† Can be doubled using the 1080p capacity upgrade feature key

‡ Can be doubled using the MCU 6 to 12 HD port upgrade feature key. Can be quadrupled using both the MCU 6 to 12 HD port upgrade feature key and the 1080p capacity upgrade feature key

# Working with the event logs

If you are experiencing complex issues that require advanced troubleshooting, you may need to collect information from the MCU logs. Typically, you will be working with customer support who can help you obtain these logs.

## Event log

The last 2000 status messages generated by the MCU are displayed in the **Event log** page (**Logs > Event log**). In general these messages are provided for information, and occasionally *Warnings* or *Errors* may be shown in the Event log. The presence of such messages is not cause for concern necessarily; if you are experiencing a specific problem with the operation or performance of the MCU, Customer support can interpret logged messages and their significance for you.

You can:

- ▶ Change the level of detail collected in the traces by editing the **Capture filter** page. You should not modify these settings unless instructed to do so by customer support.
- ▶ Display the log as text: go to **Logs > Event log** and click **Download as text**.
- ▶ Change which of the stored Event log entries are displayed by editing the **Display filter** page
- ▶ Send the event log to one or more syslog servers on the network for storage or analysis. The servers are defined in the **Syslog** page. For more information, refer to [Logging using syslog](#)
- ▶ Empty the log by clicking **Clear log**.

## Event capture filter

The Event capture filter allows you to change the level of detail to collect in the Event log traces.

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**Note:** You should not modify these settings unless instructed to do so by customer support. Modifying these settings can impair the performance of your MCU.

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Normally, the capture filter should be set to the default of *Errors, warnings and information* for all logging sources. There is no advantage in changing the setting of any source without advice from customer support. There is a limited amount of space available to store logged messages and enabling anything other than *Errors, warnings and information* could cause the log to become full quickly.

## Event display filter

The Event display filter allows you to view or highlight stored Event log entries. Normally, you should not need to view or modify any of the settings on this page.

## Syslog

You can configure the MCU to send event messages to up to four syslog servers. To add or remove a syslog server, go to **Logs > Syslog** and make the changes you require. See [Logging using syslog](#).

## H.323/SIP log

The **H.323/SIP log** page records every H.323 and SIP message received or transmitted from the MCU. The log can be exported in an .xml file by clicking **Download as XML**.

By default the H.323/SIP log is disabled because it affects performance, but customer support may ask you to enable it if there is a problem with a unit in your network. To do this, click **Enable H323/SIP logging**

## Audit log

The audit log records any user action on the MCU which might compromise the security of the unit, of its functions, or of the network. For more information, refer to [Working with the audit logs](#).

## Call Detail Records

In addition to the logs described above, the MCU can also store Call Detail Records (CDR) which may be used for auditing and billing purposes. Events in the log are displayed in the CDR log page. See [Working with Call Detail Records](#) for more details.

## Understanding security warnings

The **Security status** page displays a list of active security warnings for the MCU. To access this information, go to **Status > Security**. Security warnings identify potential weaknesses in the security of the MCU's configuration. For more information on configuring security settings, refer to [Configuring security settings](#). For more detailed information on the security status, refer to [Displaying security status](#).

The table below details the warnings that appear, and the relevant actions needed to rectify them.

Warning	Action	Explanation
<b>Advanced password security is disabled</b>	Enable <b>advanced account security mode</b> in security settings	<p>If advanced account security mode is not enabled, passwords will be stored in plain text in the configuration file, and therefore be unsecure.</p> <p>To enable advanced account security mode, go to <b>Settings &gt; Security</b> and enable <i>Advanced account security mode</i>.</p>
<b>Hide log messages on console is disabled</b>	Enable hide log messages on console in serial console settings	To hide log messages on the console, go to <b>Settings &gt; Security</b> and select <b>Hide log messages on console</b> . This will stop event messages appearing on the console.
<b>Require administrator login to console is disabled</b>	Enable require administrator login in serial console settings	<p>You must log in using an admin account to access serial console commands, in this way the serial console will be more secure.</p> <p>To do this, go to <b>Settings &gt; Security</b> and select <b>Require administrator login</b>.</p>
<b>Guest account is enabled</b>	Disable the guest account.	<p>By default the guest user account is assigned the privilege of 'conference list only', meaning that users who log in as guest can view the list of active conferences and change their own profile. Disabling the guest account makes the MCU more secure.</p> <p>To disable the guest account, go to <b>Users &gt; User list</b> and select <b>Guest</b>. Select <b>Disable user account</b>.</p>
<b>Admin account has default username</b>	Change the admin account username	<p>The MCU must have at least one configured user with administrator privileges. By default, the User ID is "admin" and no password is required.</p> <p>To change the admin account username, go to <b>Users &gt; User list</b> and select <b>admin</b>. Enter a new username in the <b>User ID</b> field and click <b>Update user settings</b>.</p>
<b>Unsecured FTP service is enabled</b>	Disable FTP in network TCP services	Information sent using FTP is unencrypted and sent in plain text; therefore, it is possible for people to discover usernames and passwords easily.

Warning	Action	Explanation
		To disable FTP, go to <b>Network &gt; Services</b> and deselect the <b>FTP</b> check box.
<b>Unsecured HTTP service is enabled</b>	Disable HTTP in network TCP services	Information sent using HTTP (Web) is unsecured and not encrypted.  To disable HTTP, go to <b>Network &gt; Services</b> and deselect <b>Web</b> . We recommend that you enable <b>Secure web</b> .
<b>Unsecured SNMP service is enabled</b>	Disable SNMP in network UDP services	Information sent using SNMP is unencrypted and sent in plain text; therefore, it is possible for people to discover usernames and passwords easily.  To disable SNMP, go to <b>Network &gt; Services</b> and deselect <b>SNMP</b> .
<b>Auto-refresh of web pages is enabled</b>	Change auto-refresh interval to "No auto-refresh"	If your MCU is set to auto-refresh it could mean that on an idle MCU a session will never time out.  To turn off auto-refresh, go to <b>Settings &gt; User interface</b> and change <b>Status page auto-refresh interval</b> to <i>No auto-refresh</i> .
<b>Audit logging of configuration changes is disabled</b>	Enable the audit log	If the audit log is disabled, the MCU will not create an audit log. To enable audit logs, go to <b>Logs &gt; Audit log</b> and select <b>Enable auditing</b> .  For more information on the audit log, refer to <a href="#">Configuring security settings</a> .
<b>Audit logs dropped due to lack of compact flash, audit system integrity compromised</b>	Check system configuration for possible security changes	If no compact flash card is installed in the MCU, logs are only stored up to a maximum of 200 events. The 200 events do not 'wrap', and therefore when the maximum is reached the log is deleted and started over again. To rectify this problem, insert a compact flash card.  For more information on the audit log, refer to <a href="#">Configuring security settings</a> .
<b>Audit logs hash check failed, audit system integrity compromised</b>	Check system configuration for possible security changes	If audit logs checks fail, it is possible that your MCU has been compromised. For example, someone may have taken the compact flash card out and deleted some audit logs.  For more information on the audit log, refer to <a href="#">Configuring security settings</a>
<b>Compact flash card not present, audit and CDR logs will not be saved</b>	Insert a compact flash card or check whether the existing compact flash card is functional	If no compact flash card is installed in the MCU, logs are only stored up to a maximum of 200 events. The 200 events do not 'wrap', and therefore when the maximum is reached the log is deleted and

Warning	Action	Explanation
		<p>started over again.</p> <p>The MCU will give you this warning when you are nearing the 200 maximum. To rectify this problem, insert a compact flash card.</p>
<b>Call encryption is disabled</b>	Enable call encryption	<p>When encryption status is <i>Disabled</i>, no calls on the MCU will be able to use encryption.</p> <p>To enable encryption, go to <b>Settings &gt; Encryption</b>. For <b>Encryption status</b>, select <i>Enabled</i>.</p>
<b>Audit log above 75% capacity</b>	Download and delete audit logs	<p>The audit log has a maximum capacity of 100,000 audit events, or the size limit of the compact flash card. When you are nearing either of these limits, the MCU will give you this warning. If you reach full capacity of the compact flash card, the MCU will 'wrap' meaning that older logs will be deleted. To rectify this problem download and clear the audit log.</p> <p>To do this, go to <b>Logs &gt; Audit log</b> and select <b>Download as XML</b>. Once this has completed, click <b>Delete all records</b>.</p>
<b>Audit log above 90% capacity</b>	Download and delete audit logs.	<p>The audit log has a maximum capacity of 100,000 audit events, or the size limit of the compact flash card. When you are nearing either of these limits, the MCU will give you this warning. If you reach full capacity of the compact flash card, the MCU will 'wrap' meaning that older logs will be deleted. To rectify this problem download and clear the audit log.</p> <p>To do this, go to <b>Logs &gt; Audit log</b> and select <b>Download as XML</b>. Once this has completed, click <b>Delete all records</b>.</p>
<b>Streaming enabled</b>	Disable streaming.	<p>Streaming connections are not connected using HTTPS and are therefore less secure.</p> <p>To disable streaming, go to <b>Settings &gt; Streaming</b>. Under <b>Streaming &amp; ConferenceMe settings</b>, for <b>Enable</b> select <i>None</i>.</p>
<b>ConferenceMe enabled</b>	Disable ConferenceMe.	To disable ConferenceMe, go to <b>Settings &gt; Streaming</b> . In the <b>Streaming &amp; ConferenceMe settings</b> section, for <b>Enable</b> select <i>None</i> .
<b>Streaming enabled but streaming participants</b>	Enable streaming participants overlaid icon.	The MCU provides icons in the corner of the video screen to give participants

Warning	Action	Explanation
<b>overlaid icon disabled</b>		information about the conference.
<b>Audio participants overlaid icon disabled</b>	Enable audio participants overlaid icon.	See <a href="#">Using in-conference features with video endpoints</a> to see all in-conference icons and their descriptions.
<b>Unsecured conferences overlaid icon disabled</b>	Enable unsecured conferences overlaid icon.	To enable the icons, go to <b>Settings &gt; Conferences</b> . For <b>Overlaid icons</b> , select the icons you would like to be visible to participants.
<b>Recording indicator overlaid icon disabled</b>	Enable recording indicator overlaid icon.	
<b>Encryption not available on this device</b>	Add feature key for encryption.	To use encryption on your MCU you must have the Encryption feature key installed. To purchase this feature key, contact your reseller.
<b>Default encryption setting for new ad hoc conferences set to optional</b>	Set encryption to required in the template for new ad hoc conferences.	<p>When encryption status is <i>Enabled</i>, the MCU advertises itself as being able to use encryption and will use encryption if required to do so by an endpoint.</p> <p>To rectify this problem, go to <b>Conferences &gt; Templates &gt; Ad hoc conferences</b>. Set <b>Encryption</b>, to <i>Required</i>.</p> <p>To use encryption on your MCU you must have the Encryption feature key installed. To purchase this feature key, contact your reseller.</p>
<b>SRTP encryption disabled</b>	Enable SRTP encryption.	<p>When SRTP is disabled, the MCU will not advertise that it is able to encrypt using SRTP.</p> <p>To rectify this problem, go to <b>Settings &gt; Encryption</b>. For <b>SRTP encryption</b>, select <i>Secure transports (TLS) only</i>. This means that if encryption is used for a call, the media will only be encrypted in calls that are set up using TLS.</p>
<b>SRTP encryption enabled for all transports, including insecure transports (UDP and TCP)</b>	Enable SRTP encryption for secure transports (TLS) only.	To rectify this problem, go to <b>Settings &gt; Encryption</b> . For <b>SRTP encryption</b> , select <i>Secure transports (TLS) only</i> . This means that if encryption is used for a call, the media will only be encrypted in calls that are set up using TLS.
<b>Default encryption setting for new scheduled conferences set to optional</b>	Set encryption to required in the top level conference template.	When you (or another user) create a new conference (by choosing <b>Conferences</b> and clicking <b>Add new conference</b> ), you can set the encryption setting for the conference to be either <i>Optional</i> or

Warning	Action	Explanation
		<p><i>Required.</i></p> <p>To ensure that all new scheduled conferences use encryption, go to <b>Conferences &gt; Templates</b> and for <b>Encryption</b>, select <i>Required</i>.</p>
<b>Streaming page is public</b>	Disable public streaming page.	<p>You can allow users access to the streaming list pages without having to authenticate with the MCU. By default, these pages are accessible to users who have not logged in.</p> <p>To force users to authenticate before they can access the streaming page, go to <b>Settings &gt; User interface</b>, and in the <b>Public pages</b> section, deselect <b>Streaming</b>.</p>
<b>Conference list page is public</b>	Disable public conference list page.	<p>You can allow users access to the conference list pages without having to authenticate with the MCU. By default, these pages are accessible to users who have not logged in.</p> <p>To force users to authenticate before they can access the conference list page, go to <b>Settings &gt; User interface</b>, and in the <b>Public pages</b> section, deselect <b>Conference list</b>.</p>
<b>Shell not secured for startup</b>	Disable the serial input during startup.	<p>If <b>Disable serial input during startup</b> isn't selected, the serial console is not protected during application startup. This means users will have access to debug services in the operating system.</p> <p>To disable this, go to <b>Settings &gt; Security</b>, and select <b>Disable serial input during startup</b>.</p>

# Logging using syslog

You can send the [Event log](#) to one or more syslog servers on the network for storage or analysis.

To configure the syslog facility, go to **Logs > Syslog**.

In this section:

- ▶ [Syslog settings](#)
- ▶ [Using syslog](#)

## Syslog settings

Refer to this table for assistance when configuring Syslog settings:

Field	Field description	Usage tips
<b>Host address 1 to 4</b>	Enter the IP addresses of up to four Syslog receiver hosts.	The number of packets sent to each configured host will be displayed next to its IP address.
<b>Facility value</b>	<p>A configurable value for the purposes of identifying events from the MCU on the Syslog host. Choose from the following options:</p> <ul style="list-style-type: none"> <li>▶ <i>0 - kernel messages</i></li> <li>▶ <i>1 - user-level messages</i></li> <li>▶ <i>2 - mail system</i></li> <li>▶ <i>3 - system daemons</i></li> <li>▶ <i>4 - security/authorization messages (see Note 1)</i></li> <li>▶ <i>5 - messages generated internally by syslogd</i></li> <li>▶ <i>6 - line printer subsystem</i></li> <li>▶ <i>7 - network news subsystem</i></li> <li>▶ <i>8 - UUCP subsystem</i></li> <li>▶ <i>9 - clock daemon (see Note 2)</i></li> <li>▶ <i>10 - security/authorization messages (see Note 1)</i></li> <li>▶ <i>11 - FTP daemon</i></li> <li>▶ <i>12 - NTP subsystem</i></li> <li>▶ <i>13 - log audit (see Note 1)</i></li> <li>▶ <i>14 - log alert (see Note 1)</i></li> <li>▶ <i>15 - clock daemon (see Note 2)</i></li> <li>▶ <i>16 - local use 0 (local0)</i></li> </ul>	<p>Choose a value that you will remember as being the MCU.</p> <hr/> <p>Note: Various operating system daemons and processes have been found to utilize Facilities 4, 10, 13 and 14 for security/authorization, audit, and alert messages which seem to be similar.</p> <p>Various operating systems have been found to utilize both Facilities 9 and 15 for clock (cron/at) messages.</p> <hr/> <p>Processes and daemons that have not been explicitly assigned a Facility value may use any of the "local use" facilities (16 to 21) or they may use the "user-level" facility (1) - and these are the values that we recommend you select.</p>

Field	Field description	Usage tips
	▶ 17 - local use 1 ( <i>local1</i> )	
	▶ 18 - local use 2 ( <i>local2</i> )	
	▶ 19 - local use 3 ( <i>local3</i> )	
	▶ 20 - local use 4 ( <i>local4</i> )	
	▶ 21 - local use 5 ( <i>local5</i> )	
	▶ 22 - local use 6 ( <i>local6</i> )	
	▶ 23 - local use 7 ( <i>local7</i> )	

## Using syslog

The events that are forwarded to the syslog receiver hosts are controlled by the event log capture filter.

To define a syslog server, simply enter its IP address and then click **Update syslog settings**. The number of packets sent to each configured host is displayed next to its IP address.

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Note: Each event will have a severity indicator as follows:

- ▶ 0 - Emergency: system is unusable (unused by the MCU)
  - ▶ 1 - Alert: action must be taken immediately (unused by the MCU)
  - ▶ 2 - Critical: critical conditions (unused by the MCU)
  - ▶ 3 - Error: error conditions (used by MCU *error* events)
  - ▶ 4 - Warning: warning conditions (used by MCU *warning* events)
  - ▶ 5 - Notice: normal but significant condition (used by MCU *info* events)
  - ▶ 6 - Informational: informational messages (used by MCU *trace* events)
  - ▶ 7 - Debug: debug-level messages (used by MCU *detailed trace* events)
-

## Management event receivers

If the MCU is being managed by a remote management system, for instance TANDBERG Management Suite, information on that remote system may be shown on this page. In certain circumstances you may need to remove the link between the external system and the MCU. To do so, click **Clear**.

# SIP: Advanced

## SIP implementation

The MCU implements SIP as defined in RFC 3261. Any product wishing to establish SIP calls with the MCU should implement INVITE, ACK, BYE, and CANCEL messages along with responses from 1xx to 6xx. The MCU acts as a client and does not return 5xx and 6xx responses itself; however, proxies and other intermediaries may do so.

To use a SIP registrar in conjunction with the MCU, you must register an ID for the MCU with the SIP registrar. The MCU can register itself, and individual conferences and auto attendants with a SIP registrar.

For video Fast Update Requests, the MCU uses a type that involves sending an INFO message with an XML body. This only applies to video endpoints, but these endpoints should be able to correctly reply to INFO requests whether or not they understand them as Fast Update Requests.

## Authentication details

The username and password that you provide on the **Settings > SIP** page are the authentication details for all SIP authentication from the MCU. That is, for the SIP registrar and any SIP proxy. If you have individual conferences registered with the SIP registrar, the username will be the numeric identifier of the conference and the password will be the one entered on the **Settings > SIP** page.

## Working with Call Detail Records

The MCU can display up to 20 pages of Call Detail Records. However, the MCU is not intended to provide long-term storage of Call Detail Records. You must download the Call Detail Records and store them elsewhere.

When the CDR log is full, the oldest logs are overwritten.

To view and control the CDR log, go to **Logs > CDR log**. Refer to the tables below for details of the options available and a description of the information displayed.

- ▶ [Call Detail Record log controls](#)
- ▶ [Call Detail Record log](#)

### Call Detail Record log controls

The CDR log can contain a lot of information. The controls in this section help you to select the information for display the you find most useful. When you have finished making changes, click **Update display** to make those changes take effect. Refer to the table below for a description of the options:

Field	Field description	Usage tips
<b>Current status</b>	This field indicates whether CDR logging is enabled or disabled. Use the two buttons ( <b>Enable logging</b> and <b>Disable logging</b> ) to change status. When you enable logging, the MCU writes the CDRs to the compact flash card.	Enabling or disabling CDR logging has immediate effect. There is no need to press <b>Update display</b> after selecting one of these buttons.  Ensure there is a compact flash card available - either in the slot on the front of the unit or internally.
<b>Messages logged</b>	The current number of CDRs in the log.	
<b>Filter string</b>	Use this field to limit the scope of the displayed Call Detail Records. The filter string is not case-sensitive.	The filter string applies to the <b>Message</b> field in the log display. If a particular record has expanded details, the filter string will apply to these as well.
<b>Expand details</b>	By default, the CDR log shows only brief details of each event. When available, select from the options listed to display more details.	Selecting <i>All</i> will show the greatest amount of detail for all messages, regardless of which other options are selected.

### Call Detail Record log

This table shows the logged Call Detail Records, subject to any filtering applied (see [Call Detail Record log controls](#), above). The fields displayed and the list's associated controls are described below:

- ▶ [Downloading and clearing the log](#)
- ▶ [CDR log display](#)

## Downloading and clearing the log

The CDR log includes all stored Call Detail Records, and all available details, regardless of the current filtering and display settings. You can download all or part of the CDR log in XML format using the web interface. When you start logging, the download button shows the range of record numbers but the delete button is greyed out until the log holds a certain number of logs.

To download the CDR log, click **Download as XML** to download all the log or **Download X to Y as XML** to download a range of events. (Note that if there are a large number of logged Call Detail Records, it may take several seconds to download and display them all.)

Note: Only download CDRs when the unit is not under heavy load, otherwise performance of the unit may be impaired.

The range of logs that you can download to the web interface works in groups. Therefore you may see **Download X to Y as XML** and Y will not increase even though the log is filling up. When a threshold is reached, then Y increases. However, you always have the option to download the full log with **Download as XML**.

In addition the web interface displays a maximum of 20 pages. If the log includes more events than can be displayed on those pages, the more recent events are displayed. Therefore you may see **Download X to Y as XML** where X keeps increasing when the page is refreshed. Again you can download the full log with **Download as XML**.

To clear the CDR log, click **Delete X to Y**. This will permanently remove Call Detail Records X to Y. Due to the way the CDR log works, it may not be possible to delete all records; the button name indicates which records can be deleted. For example, if you delete the 0-399 entries, then the 400th entry appears as the first entry in this page, even if you download the full log. The download button would then show that you can download for example 400-674 (if 674 is the maximum number of entries in the log) and the delete button will be greyed out again (because it is only available when a certain number of entries are in the log).

To avoid duplicate entries when you download repeatedly, each time delete the entries that you have just downloaded.

## CDR log display

The CDR log list shows some or all of the stored records, depending on the filtering and display settings (see [Call Detail Record log controls](#)). Click on a column heading to sort by that field. Refer to the table below to understand the fields displayed in the CDR log list:

Field	Field description	Usage tips
<b># (record number)</b>	The unique index number for this Call Detail Record.	
<b>Time</b>	The time at which the Call Detail Record was created.	Records are created as different conference events occur. The time the record was created is the time that the event occurred.  Changing the time or NTP settings of the MCU will affect the displayed time for the

Field	Field description	Usage tips
		Call Detail Records. For more information, refer to <a href="#">Further information about CDR time field</a> , below.
<b>Conference</b>	The number of the conference to which this record applies	Each new conference is created with a unique numeric index. All records pertaining to a particular conference display the same conference number. This can make auditing conference events much simpler.
<b>Message</b>	The type of the Call Detail Record, and brief details, if available.	<p>The display settings allow you to display more extensive details for different record types.</p> <p>The <b>filter string</b> allows you to select for display only records where a particular word or string occurs.</p>

### Further information about CDR time field

The CDR log time stamp is stored in UTC time and not local time like the Event log, but converted to local time when displayed in the CDR log.

Changing the time and NTP's UTC Offset (on the **Settings > Time** page) will affect the CDR log time in the following ways:

- ▶ Changing the time, either changing the system time or via an NTP update will cause new CDR logs to show the new time but no change will be made to existing logged CDR events
- ▶ With NTP enabled, setting a UTC offset will change the displayed time for all the CDR events; the stored time will remain the same because it is stored in UTC and the offset is applied for display purposes
- ▶ Enabling or disabling NTP when an offset is configured will cause the display time to change for all existing events and the UTC time will change for logging future CDR events. This is because, when NTP is disabled, the current time is treated as UTC with an offset of 0

# Customizing the user interface

In this section:

- ▶ Configuring user interface settings:
  - [Controlling the auto-refreshing of status pages on the MCU](#)
  - [Controlling the display of thumbnail preview images](#)
  - [Controlling the confirmation of participant disconnections](#)
  - [Controlling the audio and visual muting of conference participants](#)
- ▶ [Controlling the availability of public pages](#)
- ▶ [Configuring welcome messages on the Login and Home pages](#)
- ▶ [Customizing voice prompts on the MCU](#)

The MCU provides you with options for customizing the voice prompts, the viewing of thumbnail previews, the text of the welcome messages and for controlling the auto-refreshing of user interface pages.

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Note: The user interface (that is the text you see on the web interface of the MCU) can be localized by Cisco or by your reseller. This type of customization is the localization of the text on the web interface and these online help pages. That is, the text has been translated into your local language. In the case where you have a localized unit, the **Use localization package** check box will be selected. For more information refer to [Customization: more information](#).

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Some localization packages are available on the [FTP site](#).

The MCU allows you to type using any character set when entering text into the web interface. For example, when naming endpoints or users, you can use any character set you require.

## Configuring user interface settings

### Controlling the auto-refreshing of status pages on the MCU

Some pages on the MCU auto-refresh to ensure that the information displayed is current. Auto-refreshing pages keep web sessions alive indefinitely meaning that an administrator login will never timeout. This may be considered to be a security weakness, and if necessary you can disable all auto-refreshing.

To control the auto-refreshing of status pages on the MCU:

1. Go to **Settings > User interface**.
2. Choose the time interval for page auto-refreshes or, to stop pages from auto-refreshing, choose **No auto-refresh**. The status pages affected by this control are as follows:
  - **Status > General**
  - **Status > Health**
  - **Status > Conferences**
  - **Conferences > Conference name > Participants**
  - **Conferences > Conference name > Statistics**
  - **Conferences > Conference name > Participant > Statistics**

- **Conferences > Conference name > Participants > Content channel**
3. Click **Apply changes**.

## Controlling the display of thumbnail preview images

To control the display of thumbnail preview images on the MCU:

1. Go to **Settings > User interface**.
2. Choose whether you want to **Show video thumbnail images** or not. This controls whether or not you will see a preview of what an endpoint sees in the conference and participants pages that can show a preview of the conference. Note that thumbnail images will not be shown for conferences where encryption is required.
3. Click **Apply changes**.

## Controlling the confirmation of participant disconnections

The default settings for the MCU allow you to disconnect, without confirmation, individual participants from a conference on the **Conference list > Conference > Participants** page. On the **Settings > User interface** page, you can configure the MCU to display an "Are you sure?" confirmation box when attempting to disconnect an individual participant. Note that all-participant disconnections always require confirmation.

To control the confirmation of individual-participant disconnections:

1. Go to **Settings > User interface**.
2. If you want attempted individual-participant disconnections to require confirmation, enable the **Confirm individual participant disconnections** option and click **Apply changes**.

## Controlling the audio and visual muting of conference participants

Using the default settings of the MCU, a conference's **Conference list > Conference > Participants** page displays controls that allow the muting of a participant's audio and/or video contribution. The **Settings > User interface** page allows you to control the presence or absence of these controls.

To control the presence or absence of audio and visual muting controls:

1. Go to **Settings > User interface**.
2. For **Participant list controls**, select the control(s) that you want to appear on the **Participant list** page.

## Controlling the availability of public pages

You can allow users access to the streaming and conference list pages without having to authenticate with the MCU. By default, these pages are accessible to users who have not logged in. However, you can disable access as follows:

1. Go to **Settings > User interface**.
2. In the **Public pages** section, enable **Streaming** and/or **Conference list** as you require.
3. Click **Apply changes**.

## Configuring welcome messages for the Login and Home pages

You can configure a message banner to appear on the Login page of the MCU. For example, some organizations might require some legal text on the login page of the MCU. You can also configure a

message banner to appear on the Home page. You can configure a separate title (maximum: 100 characters) and text (maximum: 1500 characters) for each banner. To configure the message banners:

1. Go to **Settings > User interface**.
2. In the **Welcome messages** section, enter the text you require for the titles and the text of the messages.

## Customizing voice prompts on the MCU

By default the MCU includes English voice prompts spoken by an American woman.

These prompts are used by the MCU to provide callers with information, for example: "Connecting you to your destination".

You may wish to replace these prompts with your own in order to change the wording, language or accent used. Alternative prompts may be uploaded individually using the web interface. Alternatively, a collection of voice prompts may be uploaded in one go by means of a *resource package* (see [Uploading a customization package](#)).

Some customization packages are available on the [FTP site](#).

The customization of voice prompts is controlled via the web interface. Go to **Settings > User interface**. Refer to the sections below for details of the options available and for a description of the information displayed:

- ▶ [Using default US English voice prompts](#)
- ▶ [Uploading a customization package](#)
- ▶ [Viewing the available voice prompts](#)
- ▶ [Uploading and downloading customized voice prompts](#)
- ▶ [Voice prompt specification](#)
- ▶ [Making the best possible recordings](#)

## Using default US English voice prompts

The default set of voice prompts is provided in US English and is the standard set of voice prompts supplied with the MCU. These are spoken by a female voice in Americanized English.

If your unit is using customized voice prompts and you want to return to using the default set of voice prompts:

1. Go to **Settings > User interface**.
2. In the **Select customization** section, clear **Use customized voice prompts**.
3. If your unit was provided to you as a localized unit, clear **Use localization package**.
4. Click **Apply changes**.

The default voice prompts will be applied immediately, although it may take a few seconds before everyone connected to the MCU is able to hear the new prompts.

## Uploading a customization package

It is possible to upload a collection of alternative voice prompts to the MCU with a single upload operation, using a *customization package*. Such a package may have been supplied to you by Cisco

or one of its representatives, or you may have created the package yourself (see [Downloading a customization package](#)).

To upload a package:

1. Go to **Settings > User interface**.
2. In the **Upload customization package** section, click **Browse** and locate the *.package* file on your computer.
3. Click **Upload package**.

The upload may take several seconds, depending on the size of the package file and the speed of your network connection. When the upload is complete, a status screen will be shown, displaying some or all of the individual voice prompt customizations included in the package if the upload was a success, or an error message if the upload failed for some reason.

To apply the uploaded customization package:

- ▶ In the **Select customization** section, select **Use customized voice prompts**.

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Note: If you were already using uploaded alternative voice prompts on the MCU, then these will be immediately replaced by those in the customization package. If a particular customized file is not included in the package, then any existing customization is unchanged. This allows customization sets to be built up using several different packages if required.

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## Viewing the available voice prompts

You may review the voice prompt customizations available in the table headed **Voice prompts**. The **Voice prompts** list displays all voice prompt customizations, providing details for those which have alternatives uploaded. Because these lists can be quite long, by default they are hidden. Instead, the number of customizations (files) available is shown. If any have been modified (meaning an alternative customization has been uploaded, either individually, or as part of a package), then this is indicated by an asterisk after the table name.

To expand any list to show all customizations, click **show file details**; you may subsequently hide it again by clicking **hide file details**.

In the expanded state, the table shows, for each customization, a description of the file, the standard MCU filename for the customization, and the length and date modified (uploaded) of alternative customizations present. Extra information is provided by the following symbols:

- ▶ Customizations where an alternative is available that can be individually uploaded or downloaded are indicated by two asterisks (\*\*) after their name
- ▶ Customizations where an alternative is available that cannot be uploaded or downloaded individually are indicated by one asterisk (\*) (these are files that have been provided by customer support)
- ▶ Customizations that are part of a localization package from Cisco or your reseller are indicated by a plus sign (+)

## Uploading and downloading customized voice prompts

Refer to the sections below for details of further functionality provided by the **Installed voice prompts** list:

- ▶ [Uploading individual voice prompts](#)

- ▶ [Downloading individual voice prompts](#)
- ▶ [Downloading a customization package](#)
- ▶ [Deleting customized voice prompts](#)

### **Uploading individual voice prompts**

You may upload individual voice prompts. To do this:

1. Go to **Settings > User interface**.
2. In the **Installed voice prompts** section, click **show files details** and locate the voice prompt file you require.
3. For that voice prompt, click **upload**. You may do this regardless of whether an alternative customization has already been uploaded.
4. You will be presented with a new screen, allowing you to locate and upload the customization of your choice. Click **Browse** button to locate the voice prompt file on your computer. Voice prompt files must be in the following format:
  - Microsoft WAVE (.WAV) format
  - 16kHz (16000Hz) sample rate
  - Mono
  - Uncompressed
  - Maximum 10 seconds long

If you upload a file that is not in this format, the upload may fail or the voice prompt may sound distorted when heard by users. Use an audio editing package of your choice to make any conversions required. See [Making the best possible recordings](#) for how to obtain the best possible voice prompts for your MCU customization.

Note that in addition to the 10 second length limit per prompt, there is a total length limit of four minutes for the full set of prompts. That is, if all samples were played back-to-back, it should take no more than 240 seconds.

5. When you have located the file you wish to upload, click **Upload customization**. If the upload is successful, a page displaying the size of the file uploaded will be displayed; otherwise an error will be shown. If the upload fails, check your audio file matches the specification above before contacting your support representative.

### **Downloading individual voice prompts**

You may wish to review a customization that has been previously uploaded to the MCU. To do this,

1. Go to **Settings > User interface**.
2. In the **Installed voice prompts** section, locate the voice prompt file you require.
3. For that voice prompt, right-click **download** and choose **Save Target As** (or your web browser's equivalent operation). The file will be downloaded to your computer for reference.

Only alternative customizations can be downloaded in this way; the default voice prompts may not be downloaded. In addition, only customizations uploaded as individual files may be downloaded; those uploaded as part of a package may not be downloaded.

### **Downloading a customization package**

Once you are satisfied with your customizations, you may wish to apply the entire set to another MCU. Rather than individually uploading the alternative voice prompts to each one, you may create a *customization package*.

To create a customization package containing all of the alternative voice prompts previously uploaded:

1. Go to **Settings > User interface**.
2. Click **Download package** at the bottom of the **Installed voice prompts** list. The customization package will be downloaded to your computer.

A package may only contain resources uploaded as separate files; those uploaded as part of another package may not be included. The package download option may be unavailable if no voice prompts qualify for inclusion.

## Deleting customized voice prompts

If you are dissatisfied with a voice prompt that you have uploaded to the MCU, you may delete it in the following manner:

1. Locate the voice prompt of interest in the list.
2. Click the check box to the left of the voice prompt.
3. Click **Delete selected** to remove the voice prompt.

Only alternative voice prompts may be deleted in this way; the default voice prompts cannot be deleted. If you delete an alternative customization, it will immediately revert to the default prompt, even if you have selected **Use customized voice prompts** at the top of the page.

You may wish to delete all customizations. To do this, press **Delete all**. Remember that you may revert to the default set of voice prompts without needing to delete any alternative customizations (see [Using default voice prompts](#)).

## Voice prompt specification

Below is a complete list of the voice prompts that may be customized. The default wording is shown for each prompt. You do not have to use exactly the same wordings if they are not appropriate for your needs, and are provided only as a guide.

Filename	Default wording
<b>voice_prompt_conference_already_exists</b>	I'm sorry, there is already a conference with that number
<b>voice_prompt_conference_over</b>	Your conference is now over. Goodbye
<b>voice_prompt_connect_now</b>	I'll connect you to your conference now
<b>voice_prompt_enter_conference_id</b>	Please enter the conference code now
<b>voice_prompt_enter_conference_id_or_create</b>	Please enter the conference number followed by the pound key, or press star to create a new conference
<b>voice_prompt_enter_conference_pin</b>	Please enter the security PIN for this conference now
<b>voice_prompt_enter_new_conference_id</b>	Please enter the conference number for the conference you are creating, followed by the pound key
<b>voice_prompt_enter_new_conference_pin</b>	Please enter the PIN for the conference you are creating, followed by the pound key; if you don't want a PIN, just press the pound key
<b>voice_prompt_enter_new_conference_pin_short</b>	Please enter the PIN for the conference you are creating, followed by the pound key.
<b>voice_prompt_enter_pin</b>	Please enter the security PIN

Filename	Default wording
<b>voice_prompt_fecc_usage</b>	To join a conference you may use the far end camera controls on your remote
<b>voice_prompt_first_participant</b>	You are the first participant to join the conference
<b>voice_prompt_pin_incorrect</b>	Sorry, I did not recognize that security PIN, please try again
<b>voice_prompt_starting_now</b>	Thank you, your conference is starting now
<b>voice_prompt_two_minutes</b>	Your conference is scheduled to end in two minutes
<b>voice_prompt_unknown_conference</b>	Sorry, I did not recognize that conference code, please try again
<b>voice_prompt_waiting_for_chairperson</b>	Waiting for conference chairperson
<b>voice_prompt_welcome</b>	Hello, welcome to the conferencing system

## Making the best possible recordings

There are many factors to consider when recording alternative voice prompts in order to get the best results. Below is a summary of the points to bear in mind.

### *Recording format*

It is best to make each recording with the ideal settings and hence avoid any sample-rate or resolution changes. As discussed, the ideal format is Microsoft Wave (.WAV) format, uncompressed, mono, at 16 kHz and 16-bit resolution.

If you are unable to make mono recordings, the MCU can convert stereo recordings.

### *Background noise*

It is important to minimize background noise (hiss) as much as possible. This includes ambient noises such as road noise and slamming doors etc. but also try to keep fan noise and similar to a minimum.

When played back by the MCU, samples with background noise are very apparent.

### *Consistency*

If possible, record all voice prompts in one session. This will ensure that all voice and background conditions remain constant and the recorded voice will sound similar from prompt to prompt.

### *Volume*

Record prompts using a relatively constant loudness of voice. Although it may take some trial and error, the best recordings will result from speaking loud enough that the voice is recorded loudly compared to any residual background noise, but not so loudly that it sounds distorted when played back.

# Customization: More information

There are three customization levels on the unit (for voice-prompts, web interface, help pages, and text messages):

- ▶ the factory default files that are provided in US English
- ▶ localization files that are sometimes installed by a reseller
- ▶ customized voice prompts files that can be uploaded and downloaded by you

## Precedence

For every customizable file:

1. If there is a customization file present and **Enable customized files** is selected, that file will be used.
2. Otherwise, if **Use localization package** is selected, the unit will use the localized file.
3. If 1 and 2 are not true, then the unit will use the default US English file.

## The factory default file set

The files that compose the default file set for the web interface, the voice prompts, the help pages, and text messages cannot be deleted. If you are using your own customization files or a localized unit you can return the unit to using the default file set:

To return to the defaults:

1. Go to **Settings > User interface**.
2. Ensure both **Use localization package** and **Use customized voice prompts** are cleared.

Note that the default voice prompts will be used where there is no alternative voice prompt available, even if **Use customized voice prompts** is selected.

## Localization files

In some parts of the world, units are available where the help pages, the voice prompts, the text messages, and some of the web interface are in the local language. In this case, Cisco or the reseller has uploaded a package that provides localized files to replace files in the default file set. If you have a localized unit, you are able to select to return to the default US English file set (see above).

Localization is a global change and affects all customizable files. If you have a localized unit, you cannot upload and download localized files on a file by file basis.

Some customization packages are available on the [FTP site](#).

## Customization files

Customization files for voice prompts can be recorded and uploaded by any admin user of the MCU. These files can be uploaded one by one or as a package. You can create your own package by uploading all the files you require to an MCU and then downloading them as a package. For more information, refer to [Customizing the user interface](#). A customization package does not have to include a complete set of files. Where a file name duplicates an existing uploaded voice prompt file, that file will be overwritten.

# Backing up and restoring the configuration using FTP

You can back up and restore the configuration of the MCU through its web interface. To do so, go to **Settings > Upgrade**. For more information, refer to [Upgrading and backing up the MCU](#).

You can also save the configuration of the MCU using FTP.

## To back up the configuration via FTP:

1. Ensure that **FTP** is enabled on the **Network > Services** page.
2. Connect to the MCU using an FTP client. When asked for a user name and password, enter the same ones that you use to log in to the MCU's web interface as an administrator. You will see a file called configuration.xml. This contains the complete configuration of your MCU.
3. Copy this file and store it somewhere safe.

The backup process is now complete.

## To restore the configuration using FTP:

1. Locate the copy of the configuration.xml file that you want to restore.
2. Ensure that **FTP** is enabled on the **Network > Services** page.
3. Connect to the MCU using an FTP client. When asked for a user name and password, use the same ones that use to log in to the MCU's web interface as an administrator.
4. Upload your configuration.xml file to the MCU, overwriting the existing file on the MCU's.

The restore process is now complete.

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**Note:** that the same process can be used to transfer a configuration from one MCU unit to another. However, before doing this, be sure to keep a copy of the original feature keys from the unit whose configuration is being replaced.

If you are using the configuration file to configure a duplicate unit, for example in a network where you have more than one MCU, be aware that if the original unit was configured with a static address, you will need to reconfigure the IP address on any other units on which you have used the configuration file.

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## Network connectivity testing

The Network connectivity page can be used for troubleshooting issues that arise because of problems in the network between the MCU and a remote video conferencing device being called (or a device from which a user is attempting to call the MCU).

The Network connectivity page enables you to attempt to 'ping' another device from the MCU's web interface and perform a 'traceroute' of the network path to that device. The results show whether or not you have network connectivity between the MCU and another device. You can see from which port the MCU will route to that address. For a hostname, the IP address to which it has been resolved will be displayed.

To test connectivity with a remote device, go to **Network > Connectivity**. In the text box, enter the IP address or hostname of the device to which you want to test connectivity and click **Test connectivity**.

For each successful 'ping', the time taken for the ICMP echo packet to reach the host and for the reply packet to return to the MCU is displayed in milliseconds (the round trip time). The TTL (Time To Live) value on the echo reply is also displayed.

For each intermediate host (typically routers) on the route between the MCU and the remote device, the host's IP address and the time taken to receive a response from that host is shown. Not all devices will respond to the messages sent by the MCU to analyse the route; routing entries for non-responding devices is shown as <unknown>. Some devices are known to send invalid ICMP response packets (e.g. with invalid ICMP checksums); these responses are not recognized by the MCU and therefore these hosts' entries are also shown as <unknown>.

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Note: The ping message is sent from the MCU to the IP address of the endpoint that you enter. Therefore, if the MCU has an IP route to the given IP address, regardless of whether that route lies out of port A or port B, the ping will be successful. This feature allows the MCU's IP routing configuration to be tested, and it has no security implications.

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Note: If you are unable to ping the device then check your network configuration especially any firewalls using NAT.

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## Configuring SSL certificates

If the MCU has the *Secure management (HTTPS)* or *Encryption* feature key installed, and you enable *Secure web* on the **Network > Services** page, you will be able to access the web interface of the MCU using HTTPS. The MCU has a local certificate and private key pre-installed and this will be used by default when you access the unit using HTTPS. However, we recommend that you upload your own certificate and private key to ensure security as all MCUs have identical default certificates and keys.

To upload your own certificate and key, go to **Network > SSL certificates**. Complete the fields using the table below for help and click **Upload certificate and key**. Note that you must upload a certificate and key simultaneously. After uploading a new certificate and key, you must restart the MCU.

If you have uploaded your own certificate and key, you can remove it later if necessary; to do this, click **Delete custom certificate and key**.

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Note: A certificate and key are also required if you select to use the SIP TLS service in **Network > Services**.

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The table below details the fields you see on the **Network > SSL certificates** page.

Field	Field description	Usage tips
<b>Local certificate</b>		
<b>Subject</b>	<p>The details of the business to which the certificate has been issued:</p> <ul style="list-style-type: none"> <li>▶ <b>C:</b> the country where the business is registered</li> <li>▶ <b>ST:</b> the state or province where the business is located</li> <li>▶ <b>L:</b> the locality or city where the business is located</li> <li>▶ <b>O:</b> the legal name of the business</li> <li>▶ <b>OU:</b> the organizational unit or department</li> <li>▶ <b>CN:</b> the common name for the certificate, or the domain name</li> </ul>	
<b>Issuer</b>	The details of the issuer of the certificate.	Where the certificate has been self-issued, these details will be the same as for the <b>Subject</b> .
<b>Issued</b>	The date on which the certificate was issued.	
<b>Expires</b>	The date on which the certificate will expire.	
<b>Private key</b>	Whether the private key matches the certificate.	Your web browser uses the SSL certificate's public key to encrypt the data that it sends

Field	Field description	Usage tips
		back to the MCU. The private key is used by the MCU to decrypt that data. If the <b>Private key</b> field shows 'Key matches certificate' then the data is securely encrypted in both directions.
<b>Local certificate configuration</b>		
<b>Certificate</b>	If your organization has bought a certificate, or you have your own way of generating certificates, you can upload it. Browse to find the certificate file.	
<b>Private key</b>	Browse to find the private key file that accompanies your certificate.	
<b>Private key encryption password</b>	If your private key is stored in an encrypted format, you must enter the password here so that you can upload the key to the MCU.	
<b>Trust store</b>		
<b>Subject</b>	The details of the business to which the trust store certificate has been issued: <ul style="list-style-type: none"> <li>▶ <b>C</b>: the country where the business is registered</li> <li>▶ <b>ST</b>: the state or province where the business is located</li> <li>▶ <b>L</b>: the locality or city where the business is located</li> <li>▶ <b>O</b>: the legal name of the business</li> <li>▶ <b>OU</b>: the organizational unit or department</li> <li>▶ <b>CN</b>: the common name for the certificate, or the domain name</li> </ul>	
<b>Issuer</b>	The details of the issuer of the trust store certificate.	Where the certificate has been self-issued, these details will be the same as for the <b>Subject</b> .
<b>Issued</b>	The date on which the trust store certificate was issued.	
<b>Expires</b>	The date on which the trust store certificate will expire.	
<b>Trust store</b>	You can upload a 'trust store' of certificates that the MCU will use to verify the identity of the other end of a TLS connection.  If you have a trust store certificate on the MCU, you can delete it; to do so, click <b>Delete trust store</b> .  The trust store must be in '.pem' format.	Note that uploading a new trust store replaces the existing store.
<b>Certificate verification</b>	Choose to what extent the MCU will verify the identity	The trust store

Field	Field description	Usage tips
<b>settings</b>	<p>of the far end for a connection:</p> <ul style="list-style-type: none"><li>▶ <i>No verification</i>: all outgoing connections are permitted to proceed, even if the far end does not present a valid and trusted certificate.</li><li>▶ <i>Outgoing connections only</i>: outgoing connections are only permitted if the far end has a certificate which is trusted.</li><li>▶ <i>Outgoing connections and incoming calls</i>: outgoing connections and incoming connections for SIP calls using TLS must have a certificate which is trusted otherwise the MCU will not allow the connection to proceed.</li></ul>	<p>contains 'master' certificates that can be used to verify the identity of a certificate presented by the far end.</p> <p>Outgoing connections are connections such as SIP calls which use TLS.</p>

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