



Configuring Interoperability with Cisco Unified CCX

Last Updated: October 28, 2008

This chapter describes features in Cisco Unified Communications Manager Express (Cisco Unified CME) that provide support for interoperability between Cisco Unified CME and external feature services, such as Cisco Customer Response Solutions (CRS) with Cisco Unified Contact Center Express (Cisco Unified CCX).



Note

To configure support for computer-based CSTA client applications, such as a Microsoft Office Communicator (MOC) client or an application developed by using the Cisco Unified CME CTI SDK, see [“Configuring CTI CSTA Protocol Suite” on page 1165](#).

Finding Feature Information in This Module

Your Cisco Unified CME version may not support all of the features documented in this module. For a list of the versions in which each feature is supported, see the [“Feature Information for Interoperability Feature” section on page 1164](#).

Contents

- [Information About Interoperability with Cisco Unified CCX, page 1142](#)
- [How to Configure Interoperability with Cisco Unified CCX, page 1144](#)
- [Configuration Examples for Interoperability with Cisco Unified CCX, page 1153](#)
- [Where to Go Next, page 1162](#)
- [Additional References, page 1163](#)
- [Feature Information for Interoperability with Cisco Unified CCX, page 1164](#)

Information About Interoperability with Cisco Unified CCX

Cisco Unified CME 4.2 and later versions support interoperability between Cisco Unified CME and Cisco Customer Response Solutions (CRS) with Cisco Unified Call Center Express (Cisco Unified CCX), including enhanced call processing, device and call monitoring, unattended call transfers to multiple call center agents and basic extension mobility, and IP IVR applications.

The Cisco Unified CCX application uses the CRS platform to provide a multimedia (voice, data, and web). Cisco IP IVR functionality is available with Cisco Unified CCX and includes prompt-and-collect and call treatment.

The following functions are provided in Cisco Unified CME 4.2 and later versions:

- Support of Cisco Unified CCX Cisco Agent Desktop for use with Cisco Unified CME
- Configuration query and update between Cisco Unified CCX and Cisco Unified CME.
- SIP-based simple and supplementary call control services including:
 - Call routing between Cisco Unified CME and Cisco Unified CCX using SIP-based route point
 - First-party call control for SIP-based simple and supplementary calls
 - Call monitoring and device monitoring based on SIP presence and dialog event package
- Cisco Unified CCX session management of Cisco Unified CME
- Cisco Unified CCX device and call monitoring of agent lines and call activities in Cisco Unified CME

Provisioning and configuration information in Cisco Unified CCX is automatically provided to Cisco Unified CME. If the configuration from Cisco Unified CCX is deleted or must be modified, you can configure the same information in Cisco Unified CME by using Cisco IOS commands.

For first party call control, a route point for Cisco CRS is a peer device to Cisco Unified CME through a SIP trunk. An incoming call to Cisco Unified CME that is targeted to a call center phone is routed to Cisco Unified CCX through the route point. The call is placed in a queue and redirected to the most suitable agent by Cisco Unified CCX.

Supplementary services such as call hold, blind transfer, and semi-attended transfer are initiated by Cisco Unified CCX. Existing SIP-based simple and supplementary service call flow applies except for blind transfers. For blind transfers with Cisco Unified CCX as the transferrer, Cisco Unified CCX will stay in the active state until the transfer target answers. It drops out only after the transferred call is successfully answered. If the transfer target does not answer when ringing times out, the call is pulled back by Cisco Unified CCX and rerouted to another agent. This mechanism also applies when the transfer target is configured with call-forward all or forward no-answer. The forward configuration is ignored during blind transfer.

When a call moves between Cisco Unified CCX and Cisco Unified CME because of redirect, transfer, and conference, the SIP Call-ID continues to change. For call control purposes, Cisco Unified CME issues a unique Global Call ID (Gcid) for every outbound call leg. A Gcid remains the same for all legs of the same call in the system, and is valid for redirect, transfer, and conference events, including 3-party conferencing when a call center phone acts as a conference host.

Before Cisco IOS Release 12.4(11)XW6, if the call monitoring module in Cisco Unified CME 4.2 detected a call associated with a non default session application, such as B-ACD or a TCL script, the module was globally disabled. After the module was disabled, Cisco Unified CCX administration had to manually re-enable the call monitoring module after the session completes.

In Cisco IOS Release 12.4(11)XW6 and later releases, the call monitoring module in Cisco Unified CME does not monitor a call associated with a non default session application, such as B-ACD or a TCL script, including all calls merged into this call by way of consult transfer and conference. The module is not disabled and continues to monitor other calls.

Table 53 contains a list of tasks required to enable operability between Cisco Unified CME and Cisco Unified CCX, presented in the order in which the tasks are to be completed. This section contains information about performing tasks in the first 2 steps in this table and procedures for completing step 3.

For configuration information, see the “How to Configure Interoperability with Cisco Unified CCX” section on page 1144.

Table 53 **Tasks to Configure Interoperability between Cisco CRS and Cisco Unified CME**

Step	Task	Name of Document
1	Verify that the appropriate Cisco Unified Communications Manager Express (Cisco Unified CME) version is installed on the router. For compatibility information, see the <i>Cisco Unified Contact Center Express (Cisco Unified CCX) Software and Hardware Compatibility Guide</i> .	—
2	Configure the Cisco Unified CME router. Tip Note the XML user ID and password in Cisco Unified CME and router’s IP address.	Prerequisites, page 1144
3	Configure Cisco Unified CME to enable interoperability with Cisco Unified CCX.	How to Configure Interoperability with Cisco Unified CCX, page 1144
4	Install Cisco Unified Contact Center Express (Cisco Unified CCX) for Cisco Unified CME.	Cisco CRS Installation Guide at http://www.cisco.com/en/US/products/sw/custcosw/ps1846/product_installation_guides_list.html .
5	Perform the initial setup of Cisco CRS for Cisco Unified CME. Tip When setup launches, you are asked for the XML user ID and password, known as AXL user in Cisco CRS, that you created in Cisco Unified CME. You also must enter the router IP address.	
6	Configure Cisco Unified CME telephony subsystem to enable interoperability with Cisco Unified CCX.	“Provisioning Unified CCX for Unified CME” chapter in the appropriate <i>Cisco CRS Administration Guide</i> or <i>Cisco Unified Contact Center Express Administration Guide</i> at http://www.cisco.com/en/US/products/sw/custcosw/ps1846/products_installation_and_configuration_guides_list.html .
7	Create users and assign the agent capability in Cisco CRS.	

How to Configure Interoperability with Cisco Unified CCX

This section contains the following procedures:

- [Enabling Interoperability with Cisco Unified CCX, page 1144](#) (required)
- [SCCP: Identifying Agent Directory Numbers in Cisco Unified CME for Session Manager, page 1147](#) (required)
- [Verifying Registrations and Subscriptions in Cisco Unified CME, page 1149](#) (optional)
- [Re-creating a Session Manager in Cisco Unified CME, page 1149](#) (optional)
- [Reconfiguring a Cisco CRS Route Point as a SIP Endpoint, page 1150](#) (optional)

Enabling Interoperability with Cisco Unified CCX

To configure Cisco Unified CME to enable interoperability between Cisco Unified CME and Cisco Unified CCX, perform the following steps.



Note

A single Cisco Unified CME can support multiple session managers.

Prerequisites

- Cisco Unified CME version and Cisco IOS release that is compatible with your Cisco Unified CCX version. For compatibility information, see the [Cisco Unified Contact Center Express \(Cisco Unified CCX\) Software and Hardware Compatibility Guide](#).
- XML API must be configured to create an AXL username for Cisco Unified CCX access. For configuration information, see [“Configuring the XML API” on page 1219](#).



Note

During the initial setup of Cisco CRS for Cisco Unified CME, you need the AXL username and password that was configured using the **xml user** command in telephony-service configuration mode. You also need the router IP address that was configured using the **ip source-address** command in telephony-service configuration mode.

- Agent phones to be connected in Cisco Unified CME must be configured in Cisco Unified CME. When configuring a Cisco Unified CCX agent phone, use the **keep-conference endcall** command to enable conference initiators to exit from conference calls and end the conference for the remaining parties. For configuration information, see [“Configuring Conferencing” on page 753](#).
- The Cisco Unified CME router must be configured to accept incoming presence requests. For configuration information, see [“Configuring Presence Service” on page 1003](#).
- To support Desktop Monitoring and Recording, the **service phone SpanToPCPort 1** command must be configured in telephony-service configuration mode. For configuration information, see [“SCCP: Modifying Vendor Parameters for All Phones” on page 1119](#).

Restrictions

- Maximum number of *active* Cisco Unified CCX agents supported: 50.
- Multi-Party Ad Hoc and Meet-Me Conferencing are not supported.

- The following incoming calls are supported for deployment of the interoperability feature: SIP trunk calls from another Cisco Unified CME and all calls from a PSTN trunk. Other trunks, such as H.323, are supported as usual in Cisco Unified CME, however, not for customer calls to Cisco Unified CCX.

SUMMARY STEPS

1. **enable**
2. **configure terminal**
3. **voice call send-alert**
4. **voice service voip**
5. **callmonitor**
6. **gcid**
7. **allow-connections sip to sip**
8. **no supplementary-service sip moved-temporary**
9. **no supplementary-service sip refer**
10. **sip**
11. **registrar server [expires [max sec] [min sec]**
12. **end**

DETAILED STEPS

	Command or Action	Purpose
Step 1	enable Example: Router> enable	Enables privileged EXEC mode. <ul style="list-style-type: none"> • Enter your password if prompted.
Step 2	configure terminal Example: Router# configure terminal	Enters global configuration mode.
Step 3	voice call send-alert Example: Router(config)# voice call send-alert	Enables the terminating gateway to send an alert message instead of a progress message after it receives a call setup message.
Step 4	voice service voip Example: Router(config)# voice service voip	Enters voice-service configuration mode and specifies voice-over-IP encapsulation.
Step 5	callmonitor Example: Router(config-voi-serv)# callmonitor	Enables call monitoring messaging functionality. <ul style="list-style-type: none"> • Used by Cisco Unified CCX for processing and reporting.

	Command or Action	Purpose
Step 6	<p>gcid</p> <p>Example: Router(config-voi-serv)# gcid</p>	<p>Enables Global Call-ID (Gcid) for call control purposes.</p> <ul style="list-style-type: none"> Used by Cisco Unified CCX for tracking call.
Step 7	<p>allow-connections sip to sip</p> <p>Example: Router(config-voi-serv)# allow-connections sip to sip</p>	<p>Allows connections between specific types of endpoints in a VoIP network.</p>
Step 8	<p>no supplementary-service sip moved-temporary</p> <p>Example: Router(config-voi-serv)# no supplementary-service sip moved-temporary</p>	<p>Prevents the router from sending a redirect response to the destination for call forwarding.</p>
Step 9	<p>no supplementary-service sip refer</p> <p>Example: Router(config-voi-serv)# no supplementary-service sip refer</p>	<p>Prevents the router from forwarding a REFER message to the destination for call transfers.</p>
Step 10	<p>sip</p> <p>Example: Router(config-voi-serv)# sip</p>	<p>Enters SIP configuration mode.</p>
Step 11	<p>registrar server [expires [max sec] [min sec]]</p> <p>Example: Router(config-voi-sip)# registrar server expires max 600 min 60</p>	<p>Enables SIP registrar functionality in Cisco Unified CME.</p> <ul style="list-style-type: none"> expires—(Optional) Sets the active time for an incoming registration. max sec—(Optional) Maximum time for a registration to expire, in seconds. Range: 600 to 86400. Default: 3600. Recommended value: 600. min sec—(Optional) Minimum time for a registration to expire, in seconds. Range: 60 to 3600. Default: 60.
Step 12	<p>end</p> <p>Example: Router(config-voi-serv)# end</p>	<p>Exits configuration mode and enters privileged EXEC mode.</p>

SCCP: Identifying Agent Directory Numbers in Cisco Unified CME for Session Manager

To specify which directory numbers, associated with phone lines on Cisco Unified CCX agent phones, can be managed by a session manager, perform the following steps.

Prerequisites

- Up to eight session managers must be configured in Cisco Unified CME.
- Directory numbers associated with Cisco Unified CCX agent phones must be configured in Cisco Unified CME.
 - Cisco Unified CME 4.2: Directory numbers for agent phones must be configured as dual lines to allow an agent to make two call connections at the same time using one phone line button. Note that if the second line of the dual-line directory number is busy, a transfer event between phones in the solution will fail to complete.
 - Cisco Unified CME 4.3/7.0 and later versions: We recommend that directory numbers for agent phones be configured as octal lines to help to ensure that a free line with the same directory number is available for a transfer event.
 - For configuration information, see [“Configuring Phones to Make Basic Calls” on page 139](#).

Restrictions

- Only SCCP phones can be configured as agent phones in Cisco Unified CME. The Cisco VG224 Analog Phone Gateway and analog and SIP phones are supported as usual in Cisco Unified CME, however, not as Cisco Unified CCX agent phones.
- Cisco Unified IP Phone 7931 cannot be configured as an agent phone in Cisco Unified CME. Cisco Unified IP Phone 7931s are supported as usual in Cisco Unified CME, however, not as Cisco Unified CCX agent phones.
- Shared-line appearance is not supported on agent phones. A directory number cannot be associated with more than one physical agent phone at one time.
- Overlaid lines are not supported on agent phones. More than one directory number cannot be associated with a single line button on an agent phone.
- Monitored mode for a line button is not supported on agent phones. An agent phone cannot be monitored by another phone.
- Cisco Unified CCX does not support a call event that includes a different directory number; all call events must include the primary directory number. Call transfers between phones with single-line directory numbers will cause call monitoring to fail.

SUMMARY STEPS

1. **enable**
2. **configure terminal**
3. **ephone-dn** *dn-tag*
4. **allow watch**
5. **session-server** {*session-tag*[,...*session-tag*]}

6. end

DETAILED STEPS

	Command or Action	Purpose
Step 1	<p>enable</p> <p>Example: Router> enable</p>	<p>Enables privileged EXEC mode.</p> <ul style="list-style-type: none"> Enter your password if prompted.
Step 2	<p>configure terminal</p> <p>Example: Router# configure terminal</p>	<p>Enters global configuration mode.</p>
Step 3	<p>ephone-dn dn-tag</p> <p>Example: Router(config)# ephone-dn 24</p>	<p>Enters ephone-dn configuration mode.</p> <ul style="list-style-type: none"> <i>dn-tag</i>—Unique ID of an already configured directory number. The tag number corresponds to a tag number created when this directory number was initially configured.
Step 4	<p>session-server <i>session-server-tag[,...session-server-tag]</i></p> <p>Example: Router(config-ephone-dn)# session-server 1,2,3,4,6</p>	<p>Specifies which session managers are to monitor the directory number being configured.</p> <ul style="list-style-type: none"> <i>session-server-tag</i>—Unique ID session manager, configured in Cisco Unified CCX and automatically provided to Cisco Unified CME. Range: 1 to 8. <p>Tip If you do not know the value for <i>session-server-tag</i>, we recommend using 1.</p> <ul style="list-style-type: none"> Can configure up to eight session-server-tags; individual tags must be separated by commas (,). Each directory number can be managed by up to eight session managers. Each session manager can monitor more than one directory number.
Step 5	<p>allow watch</p> <p>Example: Router(config-ephone-dn)# allow watch</p>	<p>Allows the phone line associated with this directory number to be monitored by a watcher in a presence service.</p> <ul style="list-style-type: none"> This command can also be configured in ephone-dn template configuration mode and applied to one or more phones. The ephone-dn configuration has priority over the ephone-dn template configuration.
Step 6	<p>end</p> <p>Example: Router(config-ephone-dn)# end</p>	<p>Exits configuration mode and enters privileged EXEC mode.</p>

Verifying Registrations and Subscriptions in Cisco Unified CME

Before using the system, verify registrations and subscriptions for Cisco Unified CCX endpoints.

- Step 1** Use the **show sip status registrar** command to verify whether session manager and Cisco CRS route points are registered.
- Step 2** Use the **show presence subscription summary** command to verify whether Cisco CRS route points and Cisco Unified CCX agent directory numbers are subscribed.

The following is sample output from the **show presence subscription summary** command. The first two rows show the status for two route points. The next two are for logged in agent phones.

Router# **show presence subscription summary**

```

Presence Active Subscription Records Summary: 15 subscription
Watcher                               Presentity                               SubID  Expires  SibID  Status
-----
CRScontrol@10.4.171.81                 8101@10.4.171.34                         4      3600    0      idle
CRScontrol@10.4.171.81                 8201@10.4.171.34                         8      3600    0      idle
CRScontrol@10.4.171.81                 4016@10.4.171.34                        10     3600    0      idle
CRScontrol@10.4.171.81                 4020@10.4.171.34                        12     3599    0      idle

```

Re-creating a Session Manager in Cisco Unified CME



Note

Provisioning and configuration information in Cisco Unified CCX is automatically provided to Cisco Unified CME. The following task is required only if the configuration from Cisco Unified CCX is deleted or must be modified.

To re-create a session manager in Cisco Unified CME for Cisco Unified CCX, perform the following steps.

SUMMARY STEPS

1. **enable**
2. **configure terminal**
3. **voice register session-server** *session-server-tag*
4. **register-id** *name*
5. **keepalive** *seconds*
6. **end**

DETAILED STEPS

	Command or Action	Purpose
Step 1	enable Example: Router> enable	Enables privileged EXEC mode. <ul style="list-style-type: none"> Enter your password if prompted.
Step 2	configure terminal Example: Router# configure terminal	Enters global configuration mode.
Step 3	voice register session-server <i>session-server-tag</i> Example: Router(config)# voice register session-server 1	Enters voice register session-server configuration mode to enable and configure a session manager for an external feature server, such as the Cisco Unified CCX application on a Cisco CRS system. <ul style="list-style-type: none"> Range: 1 to 8. A single Cisco Unified CME can support multiple session managers.
Step 4	register id <i>name</i> Example: Router(config-register-fs)# CRS1	(Optional) Required only if the configuration from Cisco Unified CCX is deleted or must be modified. <ul style="list-style-type: none"> <i>name</i>—String for identifying Cisco Unified CCX. Can contain 1 to 30 alphanumeric characters.
Step 5	keepalive <i>seconds</i> Example: Router(config-register-fs)# keepalive 300	(Optional) Required only if the configuration from Cisco Unified CCX is deleted or must be modified. <ul style="list-style-type: none"> Keepalive duration for registration, in seconds, after which the registration expires unless Cisco Unified CCX reregisters before the registration expiry. Range: 60 to 3600. Default: 300. <p>Note Default in Cisco Unified CCX is 120.</p>
Step 6	end Example: Router(config-register-fs)# end	Exits configuration mode and enters privileged EXEC mode.

Reconfiguring a Cisco CRS Route Point as a SIP Endpoint

**Note**

Provisioning and configuration information in Cisco Unified CCX is automatically provided to Cisco Unified CME. The following task is required only if the configuration from Cisco Unified CCX is deleted or must be modified.

To reconfigure a Cisco CRS route point as a SIP endpoint in Cisco Unified CME, perform the following steps.

Prerequisites

- Directory numbers associated with Cisco CRS route points must be configured in Cisco Unified CME. For configuration information for directory numbers associated with SIP endpoints, see [“Configuring Phones to Make Basic Calls” on page 139](#).
- Directory numbers associated with Cisco CRS route points must be enabled to be watched. For configuration information, see [“Configuring Presence Service” on page 1003](#).
- The **mode cme** command must be enabled in Cisco Unified CME.

Restrictions

- Each Cisco CRS route point can be managed by only one session manager.
- Each session manager can manage more than one Cisco CRS route point.

SUMMARY STEPS

1. **enable**
2. **configure terminal**
3. **voice register dn dn-tag**
4. **number number**
5. **session-server {session-tag[,...session-tag]}**
6. **allow watch**
7. **refer target dial-peer**
8. **exit**
9. **voice register pool pool-tag**
10. **number tag dn dn-tag**
11. **session-server session-tag**
12. **codec codec-type [bytes]**
13. **dtmf-relay sip-notify**
14. **end**

DETAILED STEPS

	Command or Action	Purpose
Step 1	enable Example: Router> enable	Enables privileged EXEC mode. <ul style="list-style-type: none"> • Enter your password if prompted.
Step 2	configure terminal Example: Router# configure terminal	Enters global configuration mode.

	Command or Action	Purpose
Step 3	<p>voice register dn <i>dn-tag</i></p> <p>Example: Router(config-register-global)# voice register dn 1</p>	Enters voice register dn configuration mode to define a directory number for a SIP phone, intercom line, voice port, or a message-waiting indicator (MWI).
Step 4	<p>number <i>number</i></p> <p>Example: Router(config-register-dn)# number 2777</p>	Defines a valid number for a directory number.
Step 5	<p>session-server <i>session-server-tag[,...session-server-tag]</i></p> <p>Example: Router(config-register-dn)# session-server 1</p>	<p>Specifies which session managers are to monitor the directory number being configured.</p> <ul style="list-style-type: none"> <i>session-server-tag</i>—Unique ID session manager, configured in Cisco Unified CCX and automatically provided to Cisco Unified CME. Range: 1 to 8. <p>Tip If you do not know the value for <i>session-server-tag</i>, we recommend using 1.</p> <ul style="list-style-type: none"> Can configure up to eight session-server-tags; individual tags must be separated by commas (,). Each directory number can be managed by up to eight session managers. Each session manager can monitor more than one directory number.
Step 6	<p>allow watch</p> <p>Example: Router(config-register-dn)# allow watch</p>	Allows the phone line associated with this directory number to be monitored by a watcher in a presence service.
Step 7	<p>refer target dial-peer</p> <p>Example: Router(config-register-dn)# refer target dial-peer</p>	<p>Enables watcher to handle SIP REFER message from this directory number.</p> <ul style="list-style-type: none"> target dial-peer—Refer To portion of message is based on address from dial peer for this directory number.
Step 8	<p>exit</p> <p>Example: Router(config-register-dn)# exit</p>	Exits configuration mode to the next highest mode in the configuration mode hierarchy.
Step 9	<p>voice register pool <i>pool-tag</i></p> <p>Example: Router(config)# voice register pool 3</p>	<p>Enters voice register pool configuration mode to set device-specific parameters for a Cisco CRS route point.</p> <ul style="list-style-type: none"> A voice register pool in Cisco Unified CCX can contain up to 10 individual SIP endpoints. Subsequent pools are created for additional SIP endpoints.
Step 10	<p>number tag dn <i>dn-tag</i></p> <p>Example: Router(config-register-pool)# number 1 dn 1</p>	Associates a directory number with the route point being configured.

	Command or Action	Purpose
Step 11	session-server <i>session-server-tag</i> Example: Router(config-register-pool)# session-server 1	Identifies session manager to be used to control the route point being configured. <ul style="list-style-type: none"> <i>session-server-tag</i>—Unique number assigned to a session manager. Range: 1 to 8. The tag number corresponds to a tag number created by using the voice register session-server command.
Step 12	codec <i>g711ulaw</i> Example: Router(config-register-pool)# codec g711ulaw	Specifies the codec for the dial peer dynamically created for the route point being configured. <ul style="list-style-type: none"> <i>codec-type</i>—g711ulaw is required for Cisco Unified CCX.
Step 13	dtmf-relay <i>sip-notify</i> Example: Router(config-register-pool)# dtmf-relay sip-notify	Specifies DTMF Relay method to be used by the route point being configured.
Step 14	end Example: Router(config-register-pool)# end	Exits configuration mode and enters privileged EXEC mode.

Configuration Examples for Interoperability with Cisco Unified CCX

The following output from the **show running-configuration** command shows the configuration on a Cisco Unified CME router that will interoperate with Cisco Unified CCX.

```

!
version 12.4
service timestamps debug datetime msec
service timestamps log datetime msec
no service password-encryption
!
hostname sb-sj3-3845-uut1
!
boot-start-marker
boot-end-marker
!
card type t1 0 2
card type t1 0 3
logging buffered 1000000
no logging console
enable password password
!
no aaa new-model
network-clock-participate wic 2
network-clock-participate wic 3
ip cef
!
!
no ip dhcp use vrf connected
!

```



```
!
voice register dn 2
  session-server 1
  number 8001
  allow watch
  refer target dial-peer
!
voice register dn 3
  session-server 1
  number 8101
  allow watch
  refer target dial-peer
!
voice register dn 11
  number 2011
  name ep-sip-1-11
  mwi
!
voice register dn 12
  number 2012
  name ep-sip-1-12
  mwi
!
voice register dn 16
  number 5016
  name rp-sip-1-16
  label SIP 511-5016
  mwi
!
voice register dn 17
  number 5017
  name rp-sip-1-17
  label SIP 511-5017
  mwi
!
voice register dn 18
  number 5018
  name rp-sip-1-18
  label SIP 511-5018
  mwi
!
voice register pool 1
  session-server 1
  number 1 dn 1
  number 2 dn 2
  number 3 dn 3
  dtmf-relay sip-notify
  codec g711ulaw
!
voice register pool 11
  id mac 1111.0711.2011
  type 7970
  number 1 dn 11
  dtmf-relay rtp-nte
  voice-class codec 1
  username 5112011 password 5112011
!
voice register pool 12
  id mac 1111.0711.2012
  type 7960
  number 1 dn 12
  dtmf-relay rtp-nte
  voice-class codec 1
  username 5112012 password 5112012
```

```

!
voice register pool 16
  id mac 0017.0EBC.1500
  type 7961GE
  number 1 dn 16
  dtmf-relay rtp-nte
  voice-class codec 1
  username rp-sip-1-16 password pool16
!
voice register pool 17
  id mac 0016.C7C5.0660
  type 7971
  number 1 dn 17
  dtmf-relay rtp-nte
  voice-class codec 1
  username rp-sip-1-17 password pool17
!
voice register pool 18
  id mac 0015.629E.825D
  type 7971
  number 1 dn 18
  dtmf-relay rtp-nte
  voice-class codec 1
  username rp-sip-1-18 password pool18
!
!
!
!
!
!
!
controller T1 0/2/0
  framing esf
  clock source internal
  linecode b8zs
  pri-group timeslots 1-4,24
!
controller T1 0/2/1
  framing esf
  clock source internal
  linecode b8zs
  pri-group timeslots 1-4,24
!
controller T1 0/3/0
  framing esf
  clock source internal
  linecode b8zs
  ds0-group 0 timeslots 1-4 type e&m-immediate-start
!
controller T1 0/3/1
  framing esf
  clock source internal
  linecode b8zs
  ds0-group 0 timeslots 1-4 type e&m-immediate-start
vlan internal allocation policy ascending
!
!
!
!
interface GigabitEthernet0/0
  ip address 209.165.201.1 255.255.255.224
  duplex auto
  speed auto
  media-type rj45

```

```
!  
interface GigabitEthernet0/1  
  ip address 192.0.2.254 255.255.255.0  
  duplex auto  
  speed auto  
  media-type rj45  
!  
interface Serial0/2/0:23  
  no ip address  
  encapsulation hdlc  
  isdn switch-type primary-5ess  
  isdn protocol-emulate network  
  isdn incoming-voice voice  
  no cdp enable  
!  
interface Serial0/2/1:23  
  no ip address  
  encapsulation hdlc  
  isdn switch-type primary-5ess  
  isdn protocol-emulate network  
  isdn incoming-voice voice  
  no cdp enable  
!  
interface Service-Engine1/0  
  ip unnumbered GigabitEthernet0/0  
  service-module ip address 209.165.202.129 255.255.255.224  
  service-module ip default-gateway 209.165.201.1  
!  
ip route 192.0.0.30 255.0.0.0 192.0.0.55  
ip route 209.165.202.129 255.255.255.224 Service-Engine1/0  
ip route 192.0.2.56 255.255.255.0 209.165.202.2  
ip route 192.0.3.74 255.255.255.0 209.165.202.3  
ip route 209.165.202.158 255.255.255.224 192.0.0.55  
!  
!  
ip http server  
ip http authentication local  
ip http path flash:  
!  
!  
ixi transport http  
  response size 64  
  no shutdown  
  request outstanding 1  
!  
ixi application cme  
  no shutdown  
!  
!  
control-plane  
!  
!  
!  
voice-port 0/0/0  
!  
voice-port 0/0/1  
!  
voice-port 0/2/0:23  
!  
voice-port 0/3/0:0  
!  
voice-port 0/1/0  
!
```

```

voice-port 0/1/1
!
voice-port 0/2/1:23
!
voice-port 0/3/1:0
!
!
!
!
!
dial-peer voice 9000 voip
description ==> This is for internal calls to CUE
destination-pattern 9...
voice-class codec 1
session protocol sipv2
session target ipv4:209.165.202.129
dtmf-relay rtp-nte sip-notify
!
dial-peer voice 9001 voip
description ==> This is for external calls to CUE
destination-pattern 5119...
voice-class codec 1
session protocol sipv2
session target ipv4:209.165.202.129
dtmf-relay rtp-nte sip-notify
!
dial-peer voice 521 voip
destination-pattern 521....
voice-class codec 1
max-redirects 5
session protocol sipv2
session target ipv4:209.165.201.2
dtmf-relay rtp-nte sip-notify
!
dial-peer voice 531 voip
destination-pattern 531....
voice-class codec 1
max-redirects 5
session protocol sipv2
session target ipv4:209.165.201.3
dtmf-relay rtp-nte sip-notify
!
!
presence
presence call-list
watcher all
allow subscribe
!
sip-ua
mwi-server ipv4:209.165.202.128 expires 3600 port 5060 transport udp
presence enable
!
!
telephony-service
no auto-reg-ephone
xml user axluser password axlpass 15 <====AXL username and password for Cisco CRS
max-ephones 240
max-dn 720
ip source-address 192.0.2.254 port 2000 <====IP address of router
system message sb-sj3-3845-uut1
url services http://192.0.2.252:6293/ipphone/jsp/sciphonexml/IPAgentInitial.jsp
url authentication http://192.0.2.252:6293/ipphone/jsp/sciphonexml/IPAgentAuthenticate.jsp
cnf-file perphone
dialplan-pattern 1 511.... extension-length 4

```

```
voicemail 9001
max-conferences 8 gain -6
call-forward pattern .T
moh flash:music-on-hold.wav
multicast moh 239.10.10.1 port 2000
transfer-system full-consult
transfer-pattern .T
create cnf-files version-stamp 7960 Jun 18 2007 07:44:25
!
!
ephone-dn 1 dual-line
  session-server 1
  number 1001
  name ag-1-1
  allow watch
  mwi sip
!
!
ephone-dn 2 dual-line
  session-server 1
  number 1002
  name ag-1-2
  allow watch
  mwi sip
!
!
ephone-dn 3 dual-line
  session-server 1
  number 1003
  name ag-1-3
  allow watch
  mwi sip
!
!
ephone-dn 4 dual-line
  session-server 1
  number 1004
  name ag-1-4
  allow watch
  mwi sip
!
!
ephone-dn 5
  session-server 1
  number 1005
  name ag-1-5
  allow watch
  mwi sip
!
!
ephone-dn 11 dual-line
  number 3011
  name ep-sccp-1-11
  mwi sip
!
!
ephone-dn 12
  number 3012
  name ep-sccp-1-12
  mwi sip
!
!
ephone-dn 16 dual-line
  number 4016
```

```

label SCCP 511-4016
name rp-sccp-1-16
mwi sip
!
!
ephone-dn 17 dual-line
number 4017
label SCCP 511-4017
name rp-sccp-1-17
mwi sip
!
!
ephone-dn 18 dual-line
number 4018
label SCCP 511-4018
name rp-sccp-1-18
mwi sip
!
!
ephone-dn 19 dual-line
number 4019
label SCCP 511-4019
name rp-sccp-1-19
mwi sip
!
!
ephone-dn 20 dual-line
number 4020
label SCCP 511-4020
name rp-sccp-1-20
mwi sip
!
!
ephone-dn 21 dual-line
number 4021
label SCCP 511-4021
name rp-sccp-1-21
mwi sip
!
!
ephone-dn 22 dual-line
number 4022
label SCCP 511-4022
name rp-sccp-1-22
mwi sip
!
!
ephone 1
mac-address 1111.0711.1001
type 7970
keep-conference endcall
button 1:1
!
!
!
ephone 2
mac-address 1111.0711.1002
type 7970
keep-conference endcall
button 1:2
!
!
!
ephone 3

```

```
mac-address 1111.0711.1003
type 7970
keep-conference endcall
button 1:3
!
!
!
ephone 4
mac-address 1111.0711.1004
type 7970
keep-conference endcall
button 1:4
!
!
!
ephone 5
mac-address 1111.0711.1005
type 7970
keep-conference endcall
button 1:5
!
!
!
ephone 11
mac-address 1111.0711.3011
type 7970
keep-conference endcall
button 1:11
!
!
!
ephone 12
mac-address 1111.0711.3012
type 7960
keep-conference endcall
button 1:12
!
!
!
ephone 16
mac-address 0012.D916.5AD6
type 7960
keep-conference endcall
button 1:16
!
!
!
ephone 17
mac-address 0013.1AA6.7A9E
type 7960
keep-conference endcall
button 1:17
!
!
!
ephone 18
mac-address 0012.80F3.B013
type 7960
keep-conference endcall
button 1:18
!
!
!
ephone 19
```

```

mac-address 0013.1A1F.6282
type 7970
keep-conference endcall
button 1:19
!
!
!
ephone 20
mac-address 0013.195A.00D0
type 7970
keep-conference endcall
button 1:20
!
!
!
ephone 21
mac-address 0017.0EBC.147C
type 7961GE
keep-conference endcall
button 1:21
!
!
!
ephone 22
mac-address 0016.C7C5.0578
type 7971
keep-conference endcall
button 1:22
!
!
!
line con 0
exec-timeout 0 0
stopbits 1
line aux 0
stopbits 1
line 66
no activation-character
no exec
transport preferred none
transport input all
transport output pad telnet rlogin lapb-ta mop udptn v120
line vty 0 4
password lab
login
!
scheduler allocate 20000 1000
!
end

```

Where to Go Next

If you are done modifying parameters for phones in Cisco Unified CME, generate a new configuration file and restart the phones. See [“Generating Configuration Files for Phones” on page 261](#).

Additional References

The following sections provide references related to Cisco Unified CME features.

Related Documents

Related Topic	Document Title
Cisco Unified CME configuration	<ul style="list-style-type: none"> • Cisco Unified CME Command Reference • Cisco Unified CME Documentation Roadmap
Cisco IOS commands	<ul style="list-style-type: none"> • Cisco IOS Voice Command Reference • Cisco IOS Software Releases 12.4T Command References
Cisco IOS configuration	<ul style="list-style-type: none"> • Cisco IOS Voice Configuration Library • Cisco IOS Software Releases 12.4T documentation
Phone documentation for Cisco Unified CME	<ul style="list-style-type: none"> • User Documentation for Cisco Unified IP Phones

Technical Assistance

Description	Link
<p>The Cisco Support website provides extensive online resources, including documentation and tools for troubleshooting and resolving technical issues with Cisco products and technologies.</p> <p>To receive security and technical information about your products, you can subscribe to various services, such as the Product Alert Tool (accessed from Field Notices), the Cisco Technical Services Newsletter, and Really Simple Syndication (RSS) Feeds.</p> <p>Access to most tools on the Cisco Support website requires a Cisco.com user ID and password.</p>	http://www.cisco.com/techsupport

Feature Information for Interoperability with Cisco Unified CCX

Table 54 lists the features in this module and enhancements to the features by version.

To determine the correct Cisco IOS release to support a specific Cisco Unified CME version, see the *Cisco Unified Communications Manager Express and Cisco IOS Software Version Compatibility Matrix* at http://www.cisco.com/en/US/docs/voice_ip_comm/cucme/requirements/guide/33matrix.htm.

Use Cisco Feature Navigator to find information about platform support and software image support. Cisco Feature Navigator enables you to determine which Cisco IOS software images support a specific software release, feature set, or platform. To access Cisco Feature Navigator, go to <http://www.cisco.com/go/cfn>. An account on Cisco.com is not required.


Note

Table 54 lists the Cisco Unified CME version that introduced support for a given feature. Unless noted otherwise, subsequent versions of Cisco Unified CME software also support that feature.

Table 54 Feature Information for Interoperability Feature

Feature Name	Cisco Unified CME Version	Modification
Interoperability with Cisco Unified CCX	4.2	Enables interoperability between Cisco Unified CME and Cisco Customer Response Solutions (CRS) 5.0 and later versions with Cisco Unified Contact Center Express (Cisco Unified CCX), including Cisco Unified IP IVR, enhanced call processing, device and call monitoring, unattended call transfers to multiple call center agents, and basic extension mobility.