



CHAPTER 12

Setting Up a Siemens Hicom 300 E (European) Analog PIMG Integration with Cisco Unity Connection

For detailed instructions for setting up a Siemens Hicom 300 E (European) analog PIMG integration with Cisco Unity Connection, see the following sections in this chapter:

- [Task List to Create the Integration with a Siemens Hicom 300 E \(European\) PIMG Phone System, page 12-1](#)
- [Requirements, page 12-2](#)
- [Programming the Siemens Hicom 300 E \(European\) PIMG Phone System for Integrating with Cisco Unity Connection, page 12-3](#)
- [Setting Up the Analog PIMG Units for a DTMF Integration, page 12-5](#)
- [Creating a New Integration with the Siemens Hicom 300 E \(European\) Phone System, page 12-19](#)

Task List to Create the Integration with a Siemens Hicom 300 E (European) PIMG Phone System

Before doing the following tasks to integrate Cisco Unity Connection with the Siemens Hicom 300 E (European) phone system by using PIMG units (media gateways), confirm that the Cisco Unity Connection server is ready for the integration by completing the applicable tasks in the *Cisco Unity Connection Installation Guide*.

1. Review the system and equipment requirements to confirm that all phone system and Cisco Unity Connection server requirements have been met. See the [“Requirements” section on page 12-2](#).
2. Plan how the voice messaging ports will be used by Cisco Unity Connection. See [Chapter 2, “Planning How the Voice Messaging Ports Will Be Used by Cisco Unity Connection.”](#)
3. Program the Siemens Hicom 300 E (European) phone system and extensions. See the [“Programming the Siemens Hicom 300 E \(European\) PIMG Phone System for Integrating with Cisco Unity Connection” section on page 12-3](#).
4. Set up the PIMG units. See the [“Setting Up the Analog PIMG Units for a DTMF Integration” section on page 12-5](#).
5. Create the integration. See the [“Creating a New Integration with the Siemens Hicom 300 E \(European\) Phone System” section on page 12-19](#).

6. Test the integration. See [Chapter 14, “Testing the Integration.”](#)
7. If this integration is a second or subsequent integration, add the applicable new user templates for the new phone system. See [Chapter 15, “Adding New User Templates for Multiple Integrations.”](#)

Requirements

The Siemens Hicom 300 E (European) integration supports configurations of the following components:

Phone System

- Siemens Hicom 300 E (European) phone system.
- Software version 3.0, release 0.17 or later.
- One or more of the applicable PIMG units. For details, see [Chapter 1, “Introduction.”](#)
- The voice messaging ports in the phone system connected by analog lines to the ports on the PIMG units.

To simplify troubleshooting, we recommend that you connect the voice messaging ports on the phone system to the ports on the PIMG units in a planned manner. For example, connect the first phone system voice messaging port to the first port on the first PIMG unit, connect the second phone system voice messaging port to the second port on the first PIMG unit, and so on. Alternatively, if you have multiple PIMG units, you can reduce answer times in the event of a PIMG unit failure by connecting the phone system ports to the PIMG units in a round-robin fashion. For example, connect the first phone system voice messaging port to the first port on the first PIMG unit, connect the second phone system voice messaging port to the first port on the second PIMG unit, and so on.

- The PIMG units connected to the same LAN or WAN that Cisco Unity Connection is connected to.
- If the PIMG units connect to a WAN, the requirements for the WAN network connections are:
 - For G.729a codec formatting, a minimum of 32.76 Kbps guaranteed bandwidth for each voice messaging port.
 - For G.711 codec formatting, a minimum of 91.56 Kbps guaranteed bandwidth for each voice messaging port.
 - No network devices that implement network address translation (NAT).
 - A maximum 200 ms one-way network latency.
- The phone system ready for the integration, as described in the documentation for the phone system.

Cisco Unity Connection Server

- Cisco Unity Connection installed and ready for the integration, as described in the *Cisco Unity Connection Installation Guide* at http://www.cisco.com/en/US/products/ps6509/prod_installation_guides_list.html.
- A license that enables the applicable number of voice messaging ports.

Centralized Voice Messaging

Cisco Unity Connection supports centralized voice messaging through the phone system, which supports various inter-phone system networking protocols including proprietary protocols such as Avaya DCS, Nortel MCDN, or Siemens CorNet, and standards-based protocols such as QSIG or DPNSS. Note that centralized voice messaging is a function of the phone system and its inter-phone system networking, not voicemail. Connection will support centralized voice messaging as long as the phone system and its inter-phone system networking are properly configured. For details, see the “Centralized Voice

Messaging” section in the “[Integrating Cisco Unity Connection with the Phone System](#)” chapter of the *Cisco Unity Design Guide Release 8.x* at http://www.cisco.com/en/US/docs/voice_ip_comm/connection/8x/design/guide/8xcucdgm.html.

Programming the Siemens Hicom 300 E (European) PIMG Phone System for Integrating with Cisco Unity Connection

The following programming instructions are provided as an example only. The specific programming for your phone system may vary depending on its configuration.



Caution

In programming the phone system, do not send calls to voice messaging ports in Cisco Unity Connection that cannot answer calls (voice messaging ports that are not set to Answer Calls). For example, if a voice messaging port is set only to Send MWI Requests, do not send calls to it.

To Program the Siemens Hicom 300 E (European) Phone System by Using the Command Line Interface

- Step 1** Use the CHA-TAPRO command to create a button table to use on the voice messaging port extensions. The button table must have the following parameters while the remaining parameters keep their default values.

Table 12-1 Button Table Parameter Settings

English		German	
Parameter	Setting	Parameter	Setting
STNO	Leave this field blank.	STNO	Leave this field blank.
STD	Enter 4 .	SNU	Enter 4 .
DIGTYP	Enter OPTIT12 .	DIGTYP	Enter OPTIT12 .
KY03	Enter MB .	KY03	Enter BK .
KY04	Enter RLS .	KY04	Enter TR .
KY07	Enter NAME .	KY07	Enter NA .
KY10	Enter CONF .	KY10	Enter KF .
KY11	Enter CONS .	KY11	Enter RF .
KY12	Enter TRANSFER .	KY12	Enter UEG .

- Step 2** If the Call Route mode on the PIMG unit will not be set to Pooled, continue to [Step 3](#). Otherwise, use the ADD-SA command to create a hunt group for the voice messaging ports that are connected to the PIMG units. The hunt group must have the following parameters set while the remaining parameters keep their default values.

Table 12-2 ADD-SA Parameter Settings for Voice Messaging Ports on the Analog PIMG Units

English		German	
Parameter	Setting	Parameter	Setting
SERVICE	Enter VCE .	DIENST	Enter SPR .
CD	Enter the pilot number for the hunt group.	RNR	Enter the pilot number for the hunt group.
DPLN	Enter 0 .	WABE	Enter 0 .
NAME	Enter name for the hunt group. For example, enter "UNITY TELEPHONSVA."	NAME	Enter name for the hunt group. For example, enter "UNITY TELEPHONSVA."
CALL PROG. STATE	Leave this field blank.	VERKEHRSSITUATION	Leave this field blank.
STNO	Enter the extensions for the ports that will answer calls. Do not include the MWI-only ports.	TLNNU	Enter the extensions for the ports that will answer calls. Do not include the MWI-only ports.
TYP	Enter LIN .	ART	Enter LIN .
CQMAX	Enter 6 .	ANOKAP	Enter 6 .
OVERFLOW	Enter - .	UEBERLAUF	Enter - .

Step 3 Use the CHA-COSSU command to enable the mailbox feature in the class of service (COS) for the voice messaging ports. The mailbox feature must have the following parameters while the remaining parameters keep their default values.

Table 12-3 HA-COSSU Parameter Settings for the Mailbox Feature

English		German	
Parameter	Setting	Parameter	Setting
TYPE	Enter COS .	ART	Enter COS .
COS	Enter the number of the class of service for which the mailbox feature will be enabled. For example, enter "47".	COS	Enter the number of the class of service for which the mailbox feature will be enabled. For example, enter "47".
VOICE	Enter TA TNOTCR MB COSXCD VCE NOANSA FWDNWK ANSYN FWDECA .	SPRACHE	Enter FBKW QVKW BRK BUC SSM RWS AULEXT ANSYN AULERU .
FAX	Enter NOCO NOTIE .		Enter NIB NQVB .
TTX	Enter NOCO NOTIE .		Enter NIB NQVB .

Table 12-3 HA-COSSU Parameter Settings for the Mailbox Feature (continued)

English		German	
VTX	Enter NOCO NOTIE.	BTX	Enter NIB NQVB.
DTE	Enter TA TNOTCR BASIC.	DEE	Enter FBKW QVKW GRUBE.

Step 4 Program each phone to forward calls to the pilot number assigned to the voice messaging ports, based on one of the Cisco Unity Connection call transfer types shown in [Table 12-4](#).

Table 12-4 Call Transfer Types

Transfer Type	Usage
Release transfer (blind transfer)	Program the phone to forward calls to the pilot number when: <ul style="list-style-type: none"> The extension is busy. The call is not answered.
Supervised transfer	Program the user station to forward calls to the pilot number only when the call is not answered (on the phone system, the number of rings before forwarding must be more than the number of rings to supervise the call). Confirm that call forwarding is disabled when the extension is busy.

Setting Up the Analog PIMG Units for a DTMF Integration

Do the following procedures to set up the analog PIMG units that are connected to the Siemens Hicom 300 E (European) phone system for a DTMF integration.

These procedures require that the following tasks have already been completed:

- The phone system is connected to the PIMG units by using analog lines.
- The PIMG units are ready to be connected to the LAN or WAN.
- The PIMG units are connected to a power source.

Fields that are not mentioned in the following procedures must keep their default values. For the default values of all fields, see the manufacturer documentation for the PIMG units.

To Download the PIMG Firmware Update Files for Analog PIMG Units

Step 1 On a Windows workstation with a high-speed Internet connection that will have access to the PIMG units, go to the Voice and Unified Communications Downloads page at <http://tools.cisco.com/support/downloads/pub/Redirect.x?mdfid=278875240>.



Note To access the software download page, you must be signed in to Cisco.com as a registered user.

This procedure describes the steps when using Internet Explorer as your web browser. If you are using a different web browser, the steps may differ.

- Step 2** In the tree control on the Downloads page, expand **Unified Communications Applications > Voice Mail and Unified Messaging > Cisco Unity**, and select **Cisco Unity Telephony Integration**.
- Step 3** On the Log In page, enter your username and password, then select **Log In**.
- Step 4** On the Select a Release page, under Latest Releases, select the most recent release.
- Step 5** In the right column, select the version of the firmware for analog PIMG units.
- Step 6** On the Download Image page, select **Download**.
- Step 7** On the Supporting Document(s) page, select **Agree**.
- Step 8** In the File Download dialog box, select **Save**.
- Step 9** In the Save As dialog box, browse to the Windows workstation that will have access the PIMG units, browse to a directory where you want to save the file, and select **Save**.
- Step 10** In the Download Complete dialog box, select **Open**. The window for extracting the PIMG firmware update files appears.
- Step 11** Select **Extract**.
- Step 12** In the Extract dialog box, browse to the directory where you want the extracted files, and select **Extract**.
- Step 13** Close the window for the extracting application.

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To Set Up the Analog PIMG Units (Firmware Version 6.x)

- Step 1** On the Windows workstation, add a temporary route to enable access to the PIMG units.
- On the Windows Start menu, select **Run**.
 - Enter **cmd**, and press **Enter**. The Command Prompt window appears.
 - At the command prompt, enter **route add 10.12.13.74 <IP Address of Workstation>**, and press **Enter**.
For example, if the IP address of the workstation is 198.1.3.25, enter “route add 10.12.13.74<space>198.1.3.25” in the Command Prompt window.
 - Close the Command Prompt window.
- Step 2** Connect a PIMG unit to the network.
- Step 3** In the web browser, go to **http://10.12.13.74**.
- Step 4** To sign in, enter the following case-sensitive settings.

Table 12-5 Sign-in Settings

Field	Settings
Username	Enter admin .
Password	Enter IpodAdmin .

- Step 5** Select **OK**.
- Step 6** On the System menu, select **Upgrade**.
- Step 7** On the Upgrade page, select **Browse**.

- Step 8** In the Choose File dialog box, browse to the directory on the Windows workstation that has the extracted PIMG firmware update files.
- Step 9** Select **Ls_<xx>.app** (where <xx> is multiple digits), and select **Open**.
- Step 10** On the Upgrade page, select **Install File**.
- Step 11** After the file is installed, a message prompting you to restart the PIMG unit appears. Select **Cancel**.



Caution Do not restart the PIMG unit until you are instructed to do so later in this procedure, even if the file installation fails. Restarting the PIMG unit at this step may prevent the PIMG unit from functioning correctly.

- Step 12** Repeat [Step 6](#) through [Step 11](#) for the following files:
- Run_<xx>FskEcho.dsp
 - Ls_<xx>.fsh
- Step 13** On the Configuration menu, select **Import/Export**.
- Step 14** On the Import/Export page, select **Browse**.
- Step 15** In the Choose File dialog box, browse to the file Ls_Cfg_Hicom300ECS.ini.
- Step 16** Select **Ls_Cfg_Hicom300ECS.ini**, and select **Open**.
- Step 17** On the Import/Export page, select **Import File**.
- Step 18** After the file is imported, a message prompting you to restart the PIMG unit appears. Select **OK**.
- Step 19** In the web browser, go to **http://10.12.13.74**.
- Step 20** To sign in, enter the following case-sensitive settings.

Table 12-6 Sign-in Settings

Field	Settings
Username	Enter admin .
Password	Enter IpodAdmin .

- Step 21** Select **OK**.
- Step 22** Do the following substeps to configure an RTP port range of 16384 to 32767.



Caution You must set the RTP port range for the PIMG units if your system uses an RTP port range of 16384 to 32767. Otherwise, Cisco Unity Connection will not be able to answer calls, and callers will hear ringing or silence.



Note The default RTP port range for PIMG units is 49000 to 50000. Some Cisco Unity Connection configurations require a different RTP port range.

- a. On the Configuration menu, select **Import/Export**.
- b. On the Import/Export page, under Export Files, select **Export All Settings**.
- c. In the File Download dialog box, select **Save**.

- d. In the Save As dialog box, browse to the Windows workstation that has access to the PIMG units, browse to a directory where you want to save the file, and select **Save**.
- e. In the Download Complete dialog box, select **Open**. Notepad opens the file Config.ini that you saved.
- f. Locate the line with the following parameter:

```
gwRTPStartPort
```
- g. Change the value of the parameter to **16384** so that the line reads as follows:

```
gwRTPStartPort = 16384
```
- h. Locate the line with the following parameter:

```
gwRTPEndPort
```
- i. Change the value of the parameter to **32767** so that the line reads as follows:

```
gwRTPEndPort = 32767
```
- j. Save the file, and exit Notepad.
- k. On the Configuration menu of the PIMG unit, select **Import/Export**.
- l. On the Import/Export page, under Browse for Import File, select **Browse**.
- m. In the Choose File dialog box, browse to the file Config.ini that you saved.
- n. Select **Config.ini**, and select **Open**.
- o. On the Import/Export page, select **Import File**.
- p. When prompted to restart the PIMG unit, select **OK**.
- q. When the PIMG unit has restarted, in the web browser, go to **http://10.12.13.74**.
- r. To sign in, enter the following case-sensitive settings.

Table 12-7 Sign-in Settings

Field	Setting
Username	Enter admin .
Password	Enter IpodAdmin .

- s. Select **OK**.

Step 23 On the System menu, select **Password**.

Step 24 On the Change Password page, enter the following settings.

Table 12-8 Change Password Page Settings

Field	Settings
Old Password	Enter IpodAdmin . (This setting is case sensitive.)

Table 12-8 Change Password Page Settings (continued)

Field	Settings
New Password	Enter your new password. (This setting is case sensitive.)
Confirm Password	Enter your new password. (This setting is case sensitive.)

- Step 25** Select **Change**.
- Step 26** On the Configuration menu, select **Mgmt Protocols**.
- Step 27** On the Management Protocols page, enter the following settings.

Table 12-9 Management Protocols Page Settings

Field	Settings
E-mail Alarms Enabled	Select No .
SNMP Traps Enabled	Select No .

- Step 28** Select **Submit**.
- Step 29** On the Configuration menu, select **Routing Table**.
- Step 30** On the Routing Table page, under Router Configuration, select **VoIP Host Groups**.
- Step 31** Under VoIP Host Groups, enter the following settings for the first VoIP Host Group.

Table 12-10 First VoIP Host Group Settings

Field	Settings
Name	Accept the default or enter another descriptive name of the VoIP host group.
Load-Balanced	(Cisco Unity Connection without a cluster) Select False . (Cisco Unity Connection with a cluster configured) Select False .
Fault-Tolerant	(Cisco Unity Connection without a cluster) Select False . (Cisco Unity Connection with a cluster configured) Select True .

- Step 32** For Cisco Unity Connection without a cluster, under Host List, enter the host name or IP address of the Cisco Unity Connection server and the server port in the format <host name or IP address>:5060.
For Cisco Unity Connection with a cluster configured, under Host List, enter the host name or IP address of the subscriber Cisco Unity Connection server (the second Cisco Unity Connection server that you installed) and the server port in the format <host name or IP address>:5060.
- Step 33** For Cisco Unity Connection without a cluster, continue to [Step 35](#). For Cisco Unity Connection with a cluster configured, select **Add Host**.
- Step 34** In the second field, enter the host name or IP address of the publisher Cisco Unity Connection server (the first Cisco Unity Connection server that you installed) and the server port in the format <host name or IP address>:5060.

**Caution**

Do not add a third host under Host List or a second host group under VoIP Host Groups. Otherwise, the Cisco Unity Connection cluster may not function correctly.

Step 35 Select **Submit**.

Step 36 Under Router Configuration, select **TDM Trunk Groups**.

Step 37 Under TDM Trunk Groups, select **Add Trunk Group**.

Step 38 Under TDM Trunk Groups, enter the following settings for the first TDM trunk group.

Table 12-11 First TDM Trunk Group Settings (Inbound Calls)

Field	Settings
Name	Enter Inbound_PBX_Calls or another unique name. This TDM trunk group will handle all incoming calls from the phone system.
Selection Direction	Select Ascending .
Selection Mode	Select Linear .
Port/Channel Content	Enter the numbers of the PIMG ports that will handle inbound calls. For example, enter “*” for all PIMG ports, or enter “1-6” for the first six PIMG ports.

Step 39 Under TDM Trunk Groups, select **Add Trunk Group**.

Step 40 Enter the following settings for the second TDM trunk group.

Table 12-12 Second TDM Trunk Group Settings (MWIs)

Field	Settings
Name	Enter MWIs or another unique name. This TDM trunk group will handle outbound MWI calls (where applicable).
Selection Direction	Select Ascending .
Selection Mode	Select Cyclic .
Port/Channel Content	Enter the numbers of the PIMG ports that will MWIs. For example, enter “*” for all PIMG ports, or enter “7,8” for the last two PIMG ports.

Step 41 Under TDM Trunk Groups, select **Add Trunk Group**.

Step 42 Enter the following settings for the third TDM trunk group.

Table 12-13 Third TDM Trunk Group Settings (Outbound Calls)

Field	Settings
Name	Enter Outbound_PBX_Calls or another unique name. This TDM trunk group will handle all outbound calls from Cisco Unity Connection.
Selection Direction	Select Descending .

Table 12-13 Third TDM Trunk Group Settings (Outbound Calls) (continued)

Field	Settings
Selection Mode	Select Linear .
Port/Channel Content	Enter * for all channels in all ports. Enter the numbers of the PIMG ports that will handle outbound (dialout) calls. For example, enter "*" for all PIMG ports, or enter "7,8" for the last two PIMG ports.

Step 43 Select **Submit**.

Step 44 Under Router Configuration, select **Inbound VoIP Rules**.

Step 45 Under Inbound VoIP Rules, uncheck the **Enabled** check box for the default rule.

Step 46 Select **Add Rule**.

Step 47 Under Inbound VoIP Rules, enter the following settings for the first new inbound VoIP rule.

Table 12-14 First New Inbound VoIP Rule Settings (MWIs)

Field	Settings
Enable	Check this check box.
Rule Label	Enter MWI_Calls or another name. This inbound VoIP rule will handle all MWI calls from Cisco Unity Connection.
Request Type	Select Message .
Originating VoIP Host Address	Enter *.

Step 48 Under Inbound VoIP Request Matching, enter the following settings.



Caution The rule that you created in [Step 47](#) must be selected. Otherwise, any changes you make will apply to another rule.

Table 12-15 Inbound VoIP Request Matching Settings

Field	Settings
Calling Number	Enter *.
Calling Name	Enter *.
Called Number	Enter *.
Called Name	Enter *.
Redirect Number	Enter *.
Redirect Name	Enter *.

Step 49 Under Outbound Routes, enter the following settings.

**Caution**

The rule that you created in [Step 47](#) must be selected. Otherwise, any changes you make will apply to another rule.

Table 12-16 Outbound Routes Settings

Field	Settings
Device Selection	
Outbound Destination	Select TDM .
Trunk Group	Select the name of the TDM trunk group that you created for MWIs in Step 40 . For example, select “MWIs.”
CPID Manipulation	
Calling Number	Enter S .
Calling Name	Enter S .
Called Number	Enter D .
Called Name	Enter D .
Redirect Number	Enter R .
Redirect Name	Enter R .
Select Primary/Alternate Route	
Primary	Select Primary .

Step 50 Under Inbound VoIP Rules, select **Add Rule**.

Step 51 Under Inbound VoIP Rules, enter the following settings for the second new inbound VoIP rule.

Table 12-17 Second New Inbound VoIP Rule Settings (Outbound Calls)

Field	Settings
Enable	Check this check box.
Rule Label	Enter Outbound_UC_Calls or another name. This inbound VoIP rule will handle all outbound calls from Cisco Unity Connection.
Request Type	Select Call .
Originating VoIP Host Address	Enter * .

Step 52 Under Inbound VoIP Request Matching, enter the following settings.

**Caution**

The rule that you created in [Step 51](#) must be selected. Otherwise, any changes you make will apply to another rule.

Table 12-18 Inbound VoIP Request Matching Settings

Field	Settings
Calling Number	Enter *.
Calling Name	Enter *.
Called Number	Enter *.
Called Name	Enter *.
Redirect Number	Enter *.
Redirect Name	Enter *.

Step 53 Under Outbound Routes, enter the following settings.



Caution The rule that you created in [Step 51](#) must be selected. Otherwise, any changes you make will apply to another rule.

Table 12-19 Outbound Routes Settings

Field	Settings
Device Selection	
Outbound Destination	Select TDM .
Trunk Group	Select the name of the TDM trunk group that you created for outbound calls in Step 42 . For example, select “Outbound_PBX_Calls.”
CPID Manipulation	
Calling Number	Enter S .
Calling Name	Enter S .
Called Number	Enter D .
Called Name	Enter D .
Redirect Number	Enter R .
Redirect Name	Enter R .
Select Primary/Alternate Route	
Primary	Select Primary .

Step 54 Select **Submit**.

Step 55 Under Router Configuration, select **Inbound TDM Rules**.

Step 56 Under Inbound TDM Rules, enter the following settings for the first inbound TDM rule.

Table 12-20 First Inbound TDM Rule Settings

Field	Settings
Enable	Check this check box.
Rule Label	Enter Inbound_Rule_1 or another name. This inbound TDM rule will handle all incoming calls from the phone system.
Request Type	Select Call .
Trunk Group	Select the name of the TDM trunk group that you created for incoming calls from the phone system in Step 38 . For example, select “Inbound_PBX_Calls.”

Step 57 Under Inbound TDM Request Matching, enter the following settings.



Caution The rule that you created in [Step 56](#) must be selected. Otherwise, any changes you make will apply to another rule.

Table 12-21 Inbound TDM Request Matching Settings

Field	Settings
Calling Number	Enter the applicable matching rule that will be used. For example, enter “*” for all.
Calling Name	Enter the applicable matching rule that will be used. For example, enter “*” for all.
Called Number	Enter the applicable matching rule that will be used. For example, enter “*” for all.
Called Name	Enter the applicable matching rule that will be used. For example, enter “*” for all.
Redirect Number	Enter the applicable matching rule that will be used. For example, enter “*” for all.
Redirect Name	Enter the applicable matching rule that will be used. For example, enter “*” for all.

Step 58 Under Outbound Routes, enter the following settings.



Caution The rule that you created in [Step 56](#) must be selected. Otherwise, any changes you make will apply to another rule.

Table 12-22 Outbound Routes Settings

Field	Settings
Device Selection	
Outbound Destination	Select VoIP .

Table 12-22 Outbound Routes Settings (continued)

Field	Settings
Host Group	Select the name of the VoIP host group that you created for Cisco Unity Connection in Step 31 .
CPID Manipulation	
Calling Number	Enter S .
Calling Name	Enter S .
Called Number	Enter D .
Called Name	Enter D .
Redirect Number	Enter R .
Redirect Name	Enter R .
Select Primary/Alternate Route	
Primary	Select Primary .

- Step 59** If you want to create more Inbound TDM rules, under Inbound TDM Rules, select **Add Rule**. Otherwise, continue to [Step 61](#).
- Step 60** Repeat [Step 56](#) through [Step 59](#) for all remaining inbound TDM rules that you want to create.
- Step 61** Select **Submit**.
- Step 62** On the Configuration menu, select **TDM > General**.
- Step 63** On the TDM General Settings page, enter the following settings.

Table 12-23 TDM General Settings Page Settings

Field	Settings
PCM Coding	Select uLaw .
Minimum Call Party Delay (ms)	Enter 500 .
Maximum Call Party Delay (ms)	Enter 2000 .
Dial Digit on Time (ms)	Enter 100 .
Dial Inter-Digit Time (ms)	Enter 100 .
Dial Pause Time (ms)	Enter 2000 .
Turn MWI On FAC	Enter the code that turns MWIs on.
Turn MWI Off FAC	Enter the code that turns MWIs off.
Outbound Call Connect Timeout (ms)	Enter 10000 .
Wait for Ringback/Connect on Blind Transfer	Select Yes .
Hunt Group Extension	Enter the pilot number of the Cisco Unity Connection voice messaging ports.

- Step 64** Select **Submit**.
- Step 65** On the Configuration menu, select **TDM > Port Enable**.
- Step 66** On the TDM Port Enabling page, select **No** for the ports that you want to disable on the PIMG unit.
- Step 67** Confirm that **Yes** is selected for all other ports on the PIMG unit.
- Step 68** Select **Submit**.
- Step 69** On the Configuration menu, select **VoIP > General**.
- Step 70** On the VoIP General Settings page, enter the following settings.

Table 12-24 VoIP General Settings Page Settings

Field	Setting
User-Agent	
Host and Domain Name	Enter the host and domain name of the PIMG unit.
Transport Type	Select UDP .
Call as Domain Name	Select No .
Invite Expiration (sec)	Enter 120 .
Server	
DNS Server Address	Enter the IP Address of the Domain Name Server that the PIMG unit will use.
Registration Server Address	Leave this field blank.
Registration Server Port	Enter 5060 .
Registration Expiration (sec)	Enter 3600 .
TCP/UDP	
UDP/TCP Transports Enabled	Select Yes .
TCP/UDP Server Port	Enter 5060 .
Proxy	
Primary Proxy Server Address	Leave this field blank.
Primary Proxy Server Port	Not applicable. Leave the default setting.
Backup Proxy Server Address	Not applicable. Leave the default setting.
Backup Proxy Server Port	Not applicable. Leave the default setting.
Proxy Query Interval	Enter 10 .
Timing	
T1 Time (ms)	Enter 400 .

Table 12-24 VoIP General Settings Page Settings (continued)



Field	Setting
T2 Time (ms)	Enter 3000 .
Monitoring	
Monitor Call Connections	Select No.

Step 71 Select **Submit**.

Step 72 On the Configuration menu, select **VoIP > Media**.

Step 73 On the VoIP Media Settings page, enter the following settings.

Table 12-25 VoIP Media Settings Page Settings

Field	Settings
Audio	
Audio Compression	Select the preferred codec for audio compression.
RTP Digit Relay Mode	Select RFC2833 .
Signaling Digit Relay Mode	Select Off .
Voice Activity Detection	Select Off .
Frame Size	Select the applicable setting: <ul style="list-style-type: none"> • G.711—20 • G.729AB—10 <div style="text-align: center;"></div> <p>Caution Failure to use the correct setting will result in recorded messages containing nothing but silence.</p>
Frames Per Packet	Select the applicable setting: <ul style="list-style-type: none"> • G.711—1 • G.729AB—2 <div style="text-align: center;"></div> <p>Caution Failure to use the correct setting will result in recorded messages containing nothing but silence.</p>

Step 74 Select **Submit**.

Step 75 On the Configuration menu, select **VoIP > QOS**.

Step 76 On the VoIP QOS Configuration page, enter the following settings.

Table 12-26 VoIP QOS Configuration Page Settings

Field	Settings
Call Control QOS Byte	Enter 104 .
RTP QOS Byte	Enter 184 .

Step 77 Select **Submit**.

Step 78 On the Configuration menu, select **IP**.

Step 79 On the IP Settings page, enter the following settings.

Table 12-27 IP Settings Page Settings

Field	Settings
Client IP Address	Enter the new IP address you want to use for the PIMG unit. (This is the IP address that you enter in Cisco Unity Connection Administration when you create the integration.)
Client Subnet Mask	Enter the new subnet mask, if the subnet mask is different from the default IP address.
Default Network Gateway Address	Enter the IP address of the default network gateway router that the PIMG units will use.
BOOTP Enabled	Select No .

Step 80 Select **Submit**.

Step 81 On the Configuration menu, select **Tone Detection**.

Step 82 On the Tone Detection page, under Call Progress Tone - Learn, in the Learn Tone Event field, select **Busy** and do the following substeps to verify that the tone is correct.

- a. From an available phone, call a second phone.
- b. Answer the second phone when it rings, and leave both handsets off so that both phones are busy.
- c. From a third phone, dial one of the busy phones.
- d. Confirm that you hear a busy tone.
- e. Hang up the third phone but leave the handsets for the other two phones off.

Step 83 Under Call Progress Tone - Learn, in the Dial String field, enter the extension that you dialed in [Step 82c](#) from the third phone.

Step 84 Select **Learn**.

Step 85 On the Tone Detection page, under Call Progress Tone - Learn, in the Learn field, select **Error** and do the following substeps to verify that the tone is correct.

- a. From an available phone, dial an extension that does not exist.
- b. Confirm that you hear the reorder or error tone.
- c. Hang up the phone.

Step 86 Under Call Progress Tone - Learn, in the Dial String field, enter the extension that you dialed in [Step 85a](#).

Step 87 Select **Learn**.

- Step 88** On the Tone Detection page, under Call Progress Tone - Learn, in the Learn field, select **Ringback** and do the following substeps to verify that the tone is correct.
- From an available phone, dial an extension that does exist
 - Confirm that you hear the ringback tone.
 - Hang up the phone.
- Step 89** Under Call Progress Tone - Learn, in the Dial String field, enter the extension that you dialed in [Step 88a](#).
- Step 90** Select **Learn**.
- Step 91** Select **Submit**.
- Step 92** Hang up the phones that you used in [Step 88](#).
- Step 93** On the System menu, select **Restart**.
- Step 94** On the Restart page, select **Restart Unit Now**.
- Step 95** Repeat [Step 2](#) through [Step 94](#) on all remaining PIMG units.
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Creating a New Integration with the Siemens Hicom 300 E (European) Phone System

After ensuring that the Siemens Hicom 300 E (European) phone system, the PIMG units, and Cisco Unity Connection are ready for the integration, do the following procedure to set up the integration and to enter the port settings.

To Create an Integration

- Step 1** Sign in to Cisco Unity Connection Administration.
- Step 2** In Cisco Unity Connection Administration, expand **Telephony Integrations**, then select **Phone System**.
- Step 3** On the Search Phone Systems page, under Display Name, select the name of the default phone system.
- Step 4** On the Phone System Basics page, in the Phone System Name field, enter the descriptive name that you want for the phone system.
- Step 5** If you want to use this phone system as the default for TRaP connections so that administrators and users without voicemail boxes can record and playback through the phone in Cisco Unity Connection web applications, check the **Default TRAP Switch** check box. If you want to use another phone system as the default for TRaP connections, uncheck this check box.
- Step 6** Select **Save**.
- Step 7** On the Phone System Basics page, in the Related Links drop-down box, select **Add Port Group** and select **Go**.
- Step 8** On the New Port Group page, enter the applicable settings and select **Save**.

Table 12-28 Settings for the New Port Group Page

Field	Setting
Phone System	Select the name of the phone system that you entered in Step 4 .
Create From	Select Port Group Template and select SIP to DMG/PIMG/TIMG in the drop-down box.
Display Name	Enter a descriptive name for the port group. You can accept the default name or enter the name that you want.
SIP Security Profile	Select 5060 .
SIP Transport Protocol	Select the SIP transport protocol that Cisco Unity Connection will use.
IPv4 Address or Host Name (<i>Cisco Unity Connection 8.5 and later</i>)	Enter the IP address of the PIMG unit that you are integrating with Cisco Unity Connection.
IPv6 Address or Host Name (<i>Cisco Unity Connection 8.5 and later</i>)	Do not enter a value in this field. IPv6 is not supported for PIMG integrations.
IP Address or Host Name (<i>Cisco Unity Connection 8.0</i>)	Enter the IP address of the PIMG unit that you are integrating with Cisco Unity Connection.
Port	Enter the SIP port of the PIMG unit that Cisco Unity Connection will connect to. We recommend that you use the default setting.  Caution This name must match the setting in the TCP/UDP Server Port field on the Configuration > VoIP > General page of the PIMG unit. Otherwise, the integration will not function correctly.

Step 9 In the Related Links drop-down box, select **Add Ports** and select **Go**.

Step 10 On the New Port page, enter the following settings and select **Save**.

Table 12-29 Settings for the New Port Page

Field	Considerations
Enabled	Check this check box.
Number of Ports	Enter 8 . (If you want to use fewer than eight voice messaging ports, enter the number of voice messaging ports that you want to use on this PIMG unit.) Note For a Cisco Unity Connection cluster, the Cisco Unity Connection server must have the number of voice messaging ports that are set up on the phone system for the PIMG integration so that this Cisco Unity Connection server can handle all voice messaging traffic for the Cisco Unity Connection cluster if one of the servers stops functioning. For example, if the phone system is set up with 16 voice messaging ports, this Cisco Unity Connection server must have 16 voice messaging ports.

Table 12-29 Settings for the New Port Page (continued)

Field	Considerations
Phone System	Select the name of the phone system that you entered in Step 4 .
Port Group	Select the name of the port group that you added in Step 8 .

Step 11 On the Search Ports page, select the display name of the first voice messaging port that you created for this phone system integration.



Note By default, the display names for the voice messaging ports are composed of the port group display name followed by incrementing numbers.

Step 12 On the Port Basics page, set the voice messaging port settings as applicable. The fields in the following table are the ones that you can change.

Table 12-30 Settings for the Voice Messaging Ports

Field	Considerations
Enabled	Check this check box to enable the port. The port is enabled during normal operation. Uncheck this check box to disable the port. When the port is disabled, calls to the port get a ringing tone but are not answered. Typically, the port is disabled only by the installer during testing.
Extension	Enter the extension for the port as assigned on the phone system.
Answer Calls	Check this check box to designate the port for answering calls. These calls can be incoming calls from unidentified callers or from users.
Perform Message Notification	Check this check box to designate the port for notifying users of messages. Assign Perform Message Notification to the least busy ports.
Send MWI Requests	Check this check box to designate the port for turning MWIs on and off. Assign Send MWI Requests to the least busy ports.
Allow TRAP Connections	Check this check box so that users can use the port for recording and playback through the phone in Cisco Unity Connection web applications. Assign Allow TRAP Connections to the least busy ports.
Outgoing Hunt Order	Enter the priority order in which Cisco Unity Connection will use the ports when dialing out (for example, if the Perform Message Notification, Send MWI Requests, or Allow TRAP Connections check box is checked). The highest numbers are used first. However, when multiple ports have the same Outgoing Hunt Order number, Cisco Unity Connection will use the port that has been idle the longest.

Step 13 Select **Save**.

Step 14 Select **Next**.

Step 15 Repeat [Step 12](#) through [Step 14](#) for all remaining voice messaging ports for the phone system.

Step 16 In Cisco Unity Connection Administration, expand **Telephony Integrations**, then select **Phone System**.

Step 17 On the Search Phone Systems page, under Display Name, select the name of the phone system that you entered in [Step 4](#).

Step 18 Repeat [Step 7](#) through [Step 17](#) for each remaining PIMG unit that will be integrated with Cisco Unity Connection.



Note Each PIMG unit is connected to one port group with the applicable voice messaging ports. For example, a system that uses five PIMG units requires five port groups, one port group for each PIMG unit.

Step 19 If another phone system integration exists, in Cisco Unity Connection Administration, expand **Telephony Integrations**, then select **Trunk**. Otherwise, skip to [Step 23](#).

Step 20 On the Search Phone System Trunks page, on the Phone System Trunk menu, select **New Phone System Trunk**.

Step 21 On the New Phone System Trunk page, enter the following settings for the phone system trunk and select **Save**.

Table 12-31 Settings for the Phone System Trunk

Field	Setting
From Phone System	Enter the display name of the phone system that you are creating a trunk for.
To Phone System	Enter the display name of the previously existing phone system that the trunk will connect to.
Trunk Access Code	Enter the extra digits that Cisco Unity Connection must dial to transfer calls through the gateway to extensions on the previously existing phone system.

Step 22 Repeat [Step 20](#) and [Step 21](#) for all remaining phone system trunks that you want to create.

Step 23 In the Related Links drop-down list, select **Check Telephony Configuration** and select **Go** to confirm the phone system integration settings.

If the test is not successful, the Task Execution Results displays one or more messages with troubleshooting steps. After correcting the problems, test the connection again.

Step 24 In the Task Execution Results window, select **Close**.