



Setting Up a Siemens Hipath 4000 Analog PIMG Integration with Cisco Unity Connection

For detailed instructions for setting up an Siemens Hipath 4000 analog PIMG integration with Cisco Unity Connection, see the following sections in this chapter:

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Task List to Create a Siemens Hipath 4000 PIMG Integration

Before doing the following tasks to integrate Unity Connection with the Siemens Hipath 4000 phone system using PIMG units (media gateways), confirm that the Unity Connection server is ready for the integration after completing the server installation, following the tasks in the “Installing Cisco Unity Connection” chapter of the *Install, Upgrade, and Maintenance Guide for Cisco Unity Connection, Release 10.x*, available at http://www.cisco.com/c/en/us/td/docs/voice_ip_comm/connection/10x/install_upgrade/guide/10xcuciumgx/10xcuciumg010.html.

1. Review the system and equipment requirements to confirm that all phone system and Unity Connection server requirements have been met. See the “[Requirements](#)” section on page 17-2.
2. Plan how the voice messaging ports are used by Unity Connection. See the “[Planning the Usage of Voice Messaging Ports in Cisco Unity Connection](#)” chapter.
3. Program the Siemens Hipath 4000 phone system and extensions. See the “[Programming Siemens Hipath 4000 Phone System for Integration](#)” section on page 17-3.
4. Set up the PIMG units.
5. Create the integration. See the “[Creating New Integration with Siemens Hipath 4000 Phone System](#)” section on page 17-20.
6. Test the integration. See the “[Testing the Integration](#)” chapter.
7. If this integration is a second or subsequent integration, add the applicable new user templates for the new phone system. See the “[Adding New User Templates for Multiple Integrations](#)” chapter.

Requirements

The Siemens Hipath 4000 integration supports configurations of the following components:

Phone System

- The Siemens Hipath 4000 phone system.
- Software version 2.0 SMR9 SMP0 or later.
- SLMA24 - Q2245 analog line card.
- One or more of the applicable PIMG units. For details, see the “Introduction” chapter.
- The voice messaging ports in the phone system connected by analog lines to the ports on the PIMG units.

We recommend that you connect the voice messaging ports on the phone system to the ports on the PIMG units in a planned manner to simplify troubleshooting. For example, the first phone system voice messaging port connects to the first port on the first PIMG unit, the second phone system voice messaging port connects to the second port on the first PIMG unit, and so on.

- The PIMG units connected to the same LAN or WAN that 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.

Unity Connection Server

- Unity Connection installed and ready for the integration after completing the server installation, following the tasks in the “Installing Cisco Unity Connection” chapter of the *Install, Upgrade, and Maintenance Guide for Cisco Unity Connection, Release 10.x*, available at http://www.cisco.com/c/en/us/td/docs/voice_ip_comm/connection/10x/install_upgrade/guide/10xcuciumgx/10xcuciumg010.html.
- A license that enables the applicable number of voice messaging ports.

Centralized Voice Messaging

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. Unity Connection supports 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 10.x with the Phone System” chapter of the *Design Guide for Cisco Unity Connection Release 10.x* at http://www.cisco.com/c/en/us/td/docs/voice_ip_comm/connection/10x/design/guide/10xcucdtx/10xcucdg050.html.

Programming Siemens Hipath 4000 Phone System for Integration

If you use programming options other than those supplied in the following procedure, the performance of the integration may be affected.

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 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 Hipath 4000 Phone System

- Step 1** In the Siemens Hipath 4000 interface, use the CHA-ZAND command for the systemwide configuration for the analog ports using the following settings.

Table 17-1 Systemwide Port Settings

Field	Setting
TYPE	Enter alldata .
TRANSFER	Enter extend .
RNGTNINT	Enter yes .
PREDIA	Enter yes .

- Step 2** Use the CHA-ZAND command for the systemwide configuration for MWIs using the following settings.

Table 17-2 Systemwide MWI Settings

Field	Setting
TYPE	Enter vmi .
MWIOOPEN	Enter yes .
ALEN	Enter the one more than the number of digits in the calling party extension. For example, if the calling party extension has four digits, enter 5.
BLEN	Enter the one more than the number of digits in the called party extension. For example, if the called party extension has four digits, enter 5.

- Step 3** Use the ADD-COSSU command to configure a class of service template that have the following classes of service for the ports that connect to the PIMG units:

For incoming lines	<ul style="list-style-type: none"> • RTAx • RTAy • TNOTCR • RKOABS • NOANSA
For outgoing lines	<ul style="list-style-type: none"> • RTAx • RTAy • TNOTCR • RKOABS • NOANSA • CDRx

Step 4 Use the ADD-SCSU command to configure each port that connects to the PIMG units using the following settings.

Table 17-3 Individual Port Settings

Field	Setting
STNO	Enter the extension of the port.
DVCFIG	Enter ANATE .
DPLN	Enter the number of digits that the dial plan uses for extensions.
ITR	Enter the number of digits that the dial plan uses for extensions.
COS1	Enter the number of the class of service that you defined in Step 3 .
COS2	Enter the number of the class of service that you defined in Step 3 .
LCOSV1	Enter the number of the class of service that you defined in Step 3 .
LCOSV2	Enter the number of the class of service that you defined in Step 3 .
LCOSD1	Enter the number of the class of service that you defined in Step 3 .
LCOSD2	Enter the number of the class of service that you defined in Step 3 .
COFIDX	Enter 0 .
SPEC	Enter SUFDIAL&SSTN .
DIAL	Enter DTMF .
DPHAR	Enter DTMFSTN&PREDIA&ASEND&BSEND .

Step 5 Use the ADD-SA command to configure a hunt group for the ports that connect to the PIMG units. The following is an example of a hunt group configuration.

```

TYPE = stn
CD = 500
DPLN = 3
CPS =
SVC = vce
STNO = 501&502&503&504&505&506&507&508
POS =

```

- Step 6** Program each phone to forward calls to the pilot number assigned to the voice messaging ports, based on one of the Unity Connection call transfer types shown in the following table.

Table 17-4 Call Transfer Types

Transfer Types	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 Hipath 4000 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 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 have access to the PIMG units, go to the Voice and Unified Communications Downloads page at <http://software.cisco.com/download/navigator.html?mdfid=280082558&i=rm>.



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 Home page, expand **Unified Communications > Unified Communications Applications > 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 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.

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 17-5 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 file **Run_<xx>.dsp**.
- Step 13** On the **System** menu, select **Upgrade**.
- Step 14** On the Upgrade page, under Import, select **Browse**.
- Step 15** In the **Choose File** dialog box, browse to the file **Ls_<xx>.fsh**.
- Step 16** Select **Ls_<xx>.fsh**, and select **Open**.
- Step 17** On the Upgrade page, select **Install File**.
- Step 18** After the file is installed, 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 17-6 Sign-in Settings

Field	Settings
Username	Enter admin .
Password	Enter IpodAdmin .

- Step 21** Select **OK**.
- Step 22** If your Unity Connection system must have an RTP port range of 16384 to 32767, do the following substeps. Otherwise, continue to [Step 23](#).

**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, Unity Connection cannot answer calls and callers hear ringing or silence.

**Note**

The default RTP port range for PIMG units is 49000 to 50000. Some 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 Settings, 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 17-7 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 17-8 Change Password Page Settings

Field	Settings
Old Password	Enter IpodAdmin . (This setting is case sensitive.)
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 17-9 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 17-10 First VoIP Host Group Settings

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

- Step 32** For Unity Connection without a cluster, under **Host List**, enter the host name or IP address of the Unity Connection server and the server port in the format <host name or IP address>:5060.
For Unity Connection with a cluster configured, under **Host List**, confirm that field contains the host name or IP address of the publisher server (the first Unity Connection server that you installed) and the server port in the format <host name or IP address>:5060.
- Step 33** For Unity Connection without a cluster, continue to [Step 36](#). For Unity Connection with a cluster configured, under **Host List**, enter the host name or IP address of the subscriber server (the second Unity Connection server that you installed) and the server port in the format <host name or IP address>:5060.
- Step 34** Select **Add Host**.
- Step 35** In the second field, enter the host name or IP address of the publisher server (the first 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 Unity Connection cluster may not function correctly.

- Step 36** Select **Submit**.
- Step 37** Under **Router Configuration**, select **TDM Trunk Groups**.
- Step 38** Under **TDM Trunk Groups**, select **Add Trunk Group**.
- Step 39** Under **TDM Trunk Groups**, enter the following settings for the first TDM trunk group.

Table 17-11 First TDM Trunk Group Settings (Inbound Calls)

Field	Settings
Name	Enter Inbound_PBX_Calls or another unique name. This TDM trunk group handles 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 handle inbound calls. For example, enter “*” for all PIMG ports, or enter “1-6” for the first six PIMG ports.

Step 40 Under **TDM Trunk Groups**, select **Add Trunk Group**.

Step 41 Enter the following settings for the second TDM trunk group.

Table 17-12 Second TDM Trunk Group Settings (MWIs)

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

Step 42 Under **TDM Trunk Groups**, select **Add Trunk Group**.

Step 43 Enter the following settings for the third TDM trunk group.

Table 17-13 Third TDM Trunk Group Settings (Outbound Calls)

Field	Settings
Name	Enter Outbound_PBX_Calls or another unique name. This TDM trunk group handles all outbound calls from Unity Connection.
Selection Direction	Select Descending .
Selection Mode	Select Linear .
Port/Channel Content	Enter * for all channels in all ports. Enter the numbers of the PIMG ports that handle outbound (dialout) calls. For example, enter “*” for all PIMG ports, or enter “7,8” for the last two PIMG ports.

Step 44 Select **Submit**.

Step 45 Under **Router Configuration**, select **Inbound VoIP Rules**.

Step 46 Under **Inbound VoIP Rules**, uncheck the **Enabled** check box for the default rule.

Step 47 Select **Add Rule**.

Step 48 Under **Inbound VoIP Rules**, enter the following settings for the first new inbound VoIP rule.

Table 17-14 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 handles all MWI calls from Unity Connection.
Request Type	Select Message .
Originating VoIP Host Address	Enter *.

Step 49 Under **Inbound VoIP Request Matching**, enter the following settings.



Caution The rule that you created in [Step 48](#) must be selected. Otherwise, any changes you make apply to another rule.

Table 17-15 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 50 Under **Outbound Routes**, enter the following settings.



Caution The rule that you created in [Step 48](#) must be selected. Otherwise, any changes you make apply to another rule.

Table 17-16 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 41 . For example, select “MWIs.”
CPID Manipulation	
Calling Number	Enter S .
Calling Name	Enter S .

Table 17-16 Outbound Routes Settings (continued)

Field	Settings
Called Number	Enter D .
Called Name	Enter D .
Redirect Number	Enter R .
Redirect Name	Enter R .
Select Primary/Alternate Route	
Primary	Select Primary .

Step 51 Under **Inbound VoIP Rules**, select **Add Rule**.

Step 52 Under **Inbound VoIP Rules**, enter the following settings for the second new inbound VoIP rule.

Table 17-17 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 handles all outbound calls from Unity Connection.
Request Type	Select Call .
Originating VoIP Host Address	Enter *.

Step 53 Under **Inbound VoIP Request Matching**, enter the following settings.



Caution The rule that you created in [Step 52](#) must be selected. Otherwise, any changes you make apply to another rule.

Table 17-18 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 54 Under **Outbound Routes**, enter the following settings.



Caution The rule that you created in [Step 52](#) must be selected. Otherwise, any changes you make apply to another rule.

Table 17-19 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 43 . 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 55 Select **Submit**.

Step 56 Under **Router Configuration**, select **Inbound TDM Rules**.

Step 57 Under **Inbound TDM Rules**, enter the following settings for the first inbound TDM rule.

Table 17-20 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 handles 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 39 . For example, select “Inbound_PBX_Calls.”

Step 58 Under **Inbound TDM Request Matching**, enter the following settings.



Caution The rule that you created in [Step 57](#) must be selected. Otherwise, any changes you make apply to another rule.

Table 17-21 Inbound TDM Request Matching Settings

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

Table 17-21 Inbound TDM Request Matching Settings (continued)

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

Step 59 Under **Outbound Routes**, enter the following settings.



Caution The rule that you created in [Step 57](#) must be selected. Otherwise, any changes you make apply to another rule.

Table 17-22 Outbound Routes Settings

Field	Settings
Device Selection	
Outbound Destination	Select VoIP .
Host Group	Select the name of the VoIP host group that you created for Unity Connection in Step 31 .
CPID Manipulation	
Calling Number	Enter the applicable matching rule used. For example, enter “*” for all.
Calling Name	Enter the applicable matching rule used. For example, enter “*” for all.
Called Number	Enter the applicable matching rule used. For example, enter “*” for all.
Called Name	Enter the applicable matching rule used. For example, enter “*” for all.
Redirect Number	Enter the applicable matching rule used. For example, enter “*” for all.
Redirect Name	Enter the applicable matching rule used. For example, enter “*” for all.
Select Primary/Alternate Route	
Primary	Select Primary .

Step 60 If you want to create more Inbound TDM rules, under **Inbound TDM Rules**, select **Add Rule**. Otherwise, continue to [Step 62](#).

Step 61 Repeat [Step 57](#) through [Step 60](#) for all remaining inbound TDM rules that you want to create.

Step 62 Select **Submit**.

Step 63 On the **Configuration** menu, select **TDM > Analog**.

Step 64 On the TDM Analog Settings page, enter the following settings.

Table 17-23 TDM Analog Settings Page Settings

Field	Settings
Timing	

Table 17-23 TDM Analog Settings Page Settings (continued)

Field	Settings

Step 65 Select **Submit**.

Step 66 On the **Configuration** menu, select **TDM > General**.

Step 67 On the TDM General Settings page, enter the following settings.

Table 17-24 TDM General Settings Page Settings

Field	Settings
PCM Coding	Select uLaw .
Minimum Call Party Delay (ms)	Enter 500 .
Maximum Call Party Delay (msecs)	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 Unity Connection voice messaging ports.
Disconnect on Fax Cleardown Tone	Select No .

Step 68 Select **Submit**.

Step 69 On the **Configuration** menu, select **TDM > Port Enable**.

Step 70 On the TDM Port Enabling page, select **No** for the ports that you want to disable on the PIMG unit.

Step 71 Confirm that **Yes** is selected for all other ports on the PIMG unit.

Step 72 Select **Submit**.

Step 73 On the **Configuration** menu, select **VoIP > General**.

Step 74 On the VoIP General Settings page, enter the following settings.

Table 17-25 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 .
SIPS URI Scheme Enabled	
Invite Expiration (sec)	Enter 120 .
Server	
DNS Server Address	Enter the IP Address of the Domain Name Server that the PIMG unit uses.
DNS Translation of Phone Numbers	
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 .
TCP Inactivity Timer (sec)	Enter 30 .
TLS	
TLS Transport Enabled	
TLS Server Port	
SSL TLS Protocol	
Mutual TLS Authentication Required	
TLS Inactivity Timer (sec)	
Verify TLS Peer Certificate Date	
Verify TLS Peer Certificate Trust	
Proxy	

Table 17-25 VoIP General Settings Page Settings (continued)

Field	Setting
Primary Proxy Server Address	Leave this field blank. Enter the IP address of the Unity Connection server.
Primary Proxy Server Port	Not applicable. Leave the default setting. For the first PIMG unit, enter 5060 . When you configure more than one PIMG unit, increase this setting by 1 for each successive unit. For example, unit 2 is 5061, unit 3 is 5062, and so on.
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 .
T2 Time (ms)	Enter 3000 .
T4 Time (ms)	Enter 5000 .
Monitoring	
Monitor Call Connections	Select No .
Call Monitor Interval (sec)	
Monitor VoIP Hosts	
VoIP Host Monitor Interval (sec)	

Step 75 Select **Submit**.



Step 76 On the **Configuration** menu, select **VoIP > Media**.

Step 77 On the VoIP Media Settings page, enter the following settings.

Table 17-26 VoIP Media Settings Page Settings

Field	Settings
Audio	
Audio Compression	Select the preferred codec for audio compression.
RTP Digit Relay Mode	Select RFC2833 .
RTP Fax/Modem Tone Relay Mode	Select RFC2833 .
RTP Source IP Address Validation	Select Off .
RTP Source UDP Port Validation	Select Off .
Signaling Digit Relay Mode	Select Off .
Voice Activity Detection	Select Off .

Table 17-26 VoIP Media Settings Page Settings (continued)

Field	Settings
RFC 3960 Early Media Support	Select On Demand .
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 results 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 results in recorded messages containing nothing but silence.</p>
Fax	
Fax IP-Transport Mode	

Step 78 Select **Submit**.

Step 79 On the **Configuration** menu, select **VoIP > QOS**.

Step 80 On the VoIP QOS Configuration page, enter the following settings.

Table 17-27 VoIP QOS Configurative Page Settings

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

Step 81 Select **Submit**.

Step 82 On the **Configuration** menu, select **IP**.

Step 83 On the IP Settings page, enter the following settings.

Table 17-28 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 use.
BOOTP Enabled	Select No .

- Step 84** Select **Submit**.
- Step 85** On the **Configure** menu, select **Tone Detection**.
- Step 86** 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.
- From an available phone, call a second phone.
 - Answer the second phone when it rings, and leave both handsets off so that both phones are busy.
 - From a third phone, dial one of the busy phones.
 - Confirm that you hear a busy tone.
 - Hang up the third phone but leave the handsets for the other two phones off.
- Step 87** Under **Call Progress Tone - Learn**, in the **Dial String** field, enter the extension that you dialed in [Step 86c](#) from the third phone.
- Step 88** Select **Learn**.
- Step 89** 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.
- From an available phone, dial an extension that does not exist.
 - Confirm that you hear the reorder or error tone.
 - Hang up the phone.
- Step 90** Under **Call Progress Tone - Learn**, in the **Dial String** field, enter the extension that you dialed in [Step 89a](#).
- Step 91** 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 92** Under **Call Progress Tone - Learn**, in the **Dial String** field, enter the extension that you dialed in [Step 91a](#).
- Step 93** Select **Learn**.
- Step 94** Select **Submit**.
- Step 95** Hang up the phones that you used in [Step 86](#).

- Step 96** On the **Configure** menu, select **Restart**.
- Step 97** On the Restart page, select **Restart Unit Now**.
- Step 98** Repeat [Step 2](#) through [Step 97](#) on all remaining PIMG units.

Creating New Integration with Siemens Hipath 4000 Phone System

After ensuring that the Siemens Hipath 4000 phone system, the PIMG units, and 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** In Cisco Unity Connection Administration, expand **Telephony Integrations**, then select **Phone System**.
- Step 2** On the Search Phone Systems page, under **Display Name**, select the name of the default phone system.
- Step 3** On the Phone System Basics page, in the **Phone System Name** field, enter the descriptive name that you want for the phone system.
- Step 4** 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 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 5** Under **Message Waiting Indicator Settings**, select **Use Same Port for Enabling and Disabling MWIs**.
- 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 17-29 Settings for the New Port Group Page

Field	Setting
Phone System	Select the name of the phone system that you entered in Step 3 .
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 Unity Connection uses.

Table 17-29 Settings for the New Port Group Page (continued)

Field	Setting
IP Address or Host Name	Enter the IP address of the PIMG unit that you are integrating with Unity Connection.
Port	Enter the SIP port of the PIMG unit that Unity Connection connects 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 cannot function correctly.

- Step 9** On the Port Group Basics page, under **Message Waiting Indicator Settings**, uncheck the **Enable Message Waiting Indicators** check box and select **Save**.
- Step 10** In the **Related Links** drop-down box, select **Add Ports** and select **Go**.
- Step 11** On the New Port page, enter the following settings and select **Save**.

Table 17-30 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 Unity Connection cluster, the server must have the number of voice messaging ports that are set up on the phone system for the PIMG integration so that this server can handle all voice messaging traffic for the cluster if one of the servers stops functioning. For example, if the phone system is set up with 16 voice messaging ports, this server must have 16 voice messaging ports.
Phone System	Select the name of the phone system that you entered in Step 3 .
Port Group	Select the name of the port group that you added in Step 8 .

- Step 12** 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 13** 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 17-31 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	<i>(Serial integrations only)</i> Uncheck this check box. Otherwise, the integration may not function correctly. <i>(Digital and analog integrations only)</i> 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 Unity Connection web applications. Assign Allow TRAP Connections to the least busy ports.
Outgoing Hunt Order	Enter the priority order in which Unity Connection uses 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, Unity Connection uses the port that has been idle the longest.

Step 14 Select **Save**.

Step 15 Select **Next**.

Step 16 Repeat [Step 13](#) through [Step 15](#) for all remaining voice messaging ports for the phone system.

Step 17 In Cisco Unity Connection Administration, expand **Telephony Integrations**, then select **Phone System**.

Step 18 On the Search Phone Systems page, under **Display Name**, select the name of the phone system that you entered in [Step 3](#).

Step 19 Repeat [Step 7](#) through [Step 18](#) for each remaining PIMG unit integrated with 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 20 To create a port group for MWIs, do the following substeps.

**Note**

All MWI requests are handled by the master PIMG unit and sent to the phone system over the RS-232 serial cable (without using voice messaging ports). So the following substeps create a separate port group without voice messaging ports and enable the port group for MWIs that are “not port specific” (they do not use ports).

- a. In Cisco Unity Connection Administration, expand **Telephony Integrations**, then select **Port Group**.
- b. On the Search Port Groups page, select **Add New**.
- c. On the New Port Group page, enter the applicable settings and select **Save**.

Table 17-32 Settings for the New Port Group Page (MWIs)

Field	Setting
Phone System	Select the name of the phone system that you entered in Step 3 .
Create From	Select Port Group Template and select SIP to DMG/PIMG/TIMG in the drop-down box.
Display Name	Enter a MWI Port Group or another descriptive name for the port group.
SIP Security Profile	Select 5060 .
SIP Transport Protocol	Select the SIP transport protocol that Unity Connection uses.
IP Address or Host Name	Enter the IP address of the master PIMG unit.
Port	Enter the SIP port of the master PIMG unit.
	 <p>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 cannot function correctly.</p>

- d. On the Port Group Basics page, on the **Edit** menu, select **Advanced Settings**.
 - e. On the Edit Advanced Settings page, under **SIP MWI Requests**, select **Not Port Specific**. and select **Save**.
 - f. On the **Edit** menu, select **Port Group Basics**.
 - g. Under **Port Group**, select **Reset**.
 - h. Under **Message Waiting Indicator Settings**, confirm that the **Enable Message Waiting Indicators** check box is checked and select **Save**.
- Step 21** If another phone system integration exists, in Cisco Unity Connection Administration, expand **Telephony Integrations**, then select **Trunk**. Otherwise, skip to [Step 25](#).
- Step 22** On the Search Phone System Trunks page, on the **Phone System Trunk** menu, select **New Phone System Trunk**.
- Step 23** On the New Phone System Trunk page, enter the following settings for the phone system trunk and select **Save**.

Table 17-33 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 connects to.
Trunk Access Code	Enter the extra digits that Unity Connection must dial to transfer calls through the gateway to extensions on the previously existing phone system.

- Step 24** Repeat [Step 22](#) and [Step 23](#) for all remaining phone system trunks that you want to create.
- Step 25** 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 26** In the **Task Execution Results** window, select **Close**.
-