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Unified Messaging Integration with CIC

Technical Reference

Abstract

This content describes configuring Customer Interaction Center when using a third-party Unified Messaging solution.

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Customer Interaction Center third-party Unified Messaging

Customer Interaction Center supports third-party solutions for Unified Messaging, such as Microsoft Exchange Server 2007 and Microsoft Exchange Server 2010. Unified Messaging centralizes a variety of communications such as voicemail, fax, and e-mail using a single repository and delivers those communications to users.

The Unified Messaging Technical Reference provides information and procedures for integrating Customer Interaction Center with a Unified Messaging solution.

Configuring third-party Unified Messaging

The third-party Unified Messaging environment is configured in Interaction Administrator. The following Interaction Administrator containers are used to support third-party Unified Messaging:

- Lines Transmissions between the Customer Interaction Center server and the Unified Messaging server are done through a Session Initiation Protocol (SIP) line.
- Stations Routing of voicemail, fax, and e-mail message between the Customer Interaction Center server and the Unified Messaging server is handled by a Unified Messaging station configured for the third-party Unified Messaging provider.
- Users Communications routing is handled by the Unified Messaging station for those users whose accounts are set to use the Unified Messaging station.

To configure third-party Unified Messaging:

- 1. Create a SIP line for Unified Messaging and configure it.
- 2. Create a Unified Messaging station and configure it.
- 3. Set User accounts to use the Unified Messaging station.
- 4. If you have multiple Unified Messaging servers available, specify and prioritize a list of proxies to provide for failover and load balancing.
- 5. Encrypt the SIP signaling stream and the Real-Time Protocol (RTP) stream to prevent malicious attacks over the connections.
- 6. Configure certificates on the Customer Interaction Center server and Unified Messaging server for connection authentication.

Creating a line

To configure a line for Unified Messaging, first create a SIP line to handle communication between the Customer Interaction Center server and the Unified Messaging server.

The following procedures are for configuring an Exchange Server 2010 Unified Messaging server:

- <u>Selecting transport settings</u>
- Disabling delayed media

Selecting transport settings

The Transport Protocol, Audio Protocol, and Receive Port must be set based on the Voice over Internet Protocol (VoIP) security setting for the Unified Messaging Dial Plan associated with the Unified Messaging server.

To select transport settings:

- 1. In the right pane of the Lines container, double-click the line to configure.
- The Line Configuration dialog box is displayed.
- 2. In the navigation box on the SIP Line Configuration page, click Transport.

ne Configuration	n			? ×
SIP Line Configura	ation Call Putback Custo	m Attributes Histor	у	
Line Audio Transport Session Authentication Proxy Registrar Headers Access Region	Transport Protocol: Audio Protocol: Address to use: Receive Port:	TCP RTP Local Area Connect Intel(R) 82567LM-3 5060	Security: Min tion Gigabit Network Con Connect Timer (ms):	imal V Inection
Recorder	Maximum Packet Retry: Maximum Invite Retry: Reinvite Delay (ms): Retryable Reason Codes:	10 6 750 480, 500-599	T1 Timer (ms): T2 Timer (ms):	500 4000
			OK Cance	el Help

3. Provide the following information:

For	Configure	
Unsecured TCP Line	Transport Protocol	ТСР
	Audio Protocol	RTP
	Receive Port	5060
Unsecured TLS Line	Transport Protocol	TLS
	Audio Protocol	RTP
	Receive Port	5061
Secured TLS Line	Transport Protocol	TLS
	Audio Protocol	SRTP
	Receive Port	5061

Disabling delayed media

Some SIP devices do not support delayed media, which may cause calls initiated from a Customer Interaction Center client to fail to connect, immediately disconnect or have one-way audio. Therefore, you need to disable delayed media for INVITE and re-INVITE requests.

To disable delayed media:

- 1. In the right pane of the Lines container, double-click the line to configure.
- The Line Configuration dialog box is displayed.
- 2. In the navigation box of the SIP Line Configuration page, click Session.

Line Configuration	n - SIP_Line_01	? ×
SIP Line Configura Line Audio Transport Session Authentication Proxy Registrar	ation Call Putback Custom Attributes History Custom Attributes History Custom Timer SIP Session Timeout: 60 seconds Disconnect on Broken RTP Media Timing: Normal	
Headers Access Region Recorder	Media reINVITE Timing: Normal Terminate Analysis on Connect Disable Media Server Passthru ASR Enabled 	
<u>«</u> » <u>c</u>	onfirm auto-save OK Cancel	Apply

3. In the Media Timing and Media reINVITE Timing boxes, select Normal.

Creating a station

To handle call routing between the Customer Interaction Center server and the Unified Messaging server, create a Unified Messaging station in Interaction Administrator.

To create a station:

- 1. Click the **Stations** container and press the **Insert** key. The **Entry Name** dialog box is displayed.
- 2. Type the station name and click **OK**.

The **Station Type** dialog is displayed.

Station Type		×
Station Type:	Remote Station Stand-alone Fax Stand-alone Phone Unified Messaging Workstation	
	Unified Messaging	Unified Messaging Provider:
Select the station template create a new station from :	that will be used for the new stations scratch.	ion. This can be left blank, if you wish to
Station Template:	ſ	
	< Back Next >	Finish Cancel Help

- 3. In the Station Type list, click Unified Messaging.
- 4. In the Unified Messaging Provider box, click Exchange.
- 5. Click Next.

The **Configuration** dialog box is displayed.

nfiguration			? >
Station Extension	: Au	to Active	
Addresses Audio Transport	Identification SIP Address:	No SIP address is configured	<u>E</u> dit
Session Authentication Region	Connection SIP Address:	No SIP address is configured	E <u>d</u> it
	Contact Line:	<use global="" setting="" sip="" station=""> 💌</use>	
	Prefix for extension (optional):		
	< <u>B</u> ack <u>N</u> ext >	Finish Cancel	Help

See the following:

- Configuring SIP addresses
- Disabling delayed media
- Adding a custom attribute

Configuring SIP addresses

For information about accessing the **Configuration** dialog box, see <u>Creating a station</u>.

1. On the Addresses page in the Configuration dialog box, click Edit to display the Identification SIP Address dialog box.

Use User Por	tion Only	
User Portion:	Host:	Port:
		506
Address:		
sip:@:5060		

2. Specify the identification SIP address to be included in the From header on INVITE messages sent by the Unified Messaging server and click **OK**.

Note: Typically, this address is either the IP address or Fully Qualified Domain Name (FQDN) of the Unified Messaging server.

3. On the Addresses page in the Configuration dialog box, click Edit to display the Connection SIP Address dialog box.

User Portion:	Host:	Port: 5060
Address:		
sib:@:2000		
Use an alternate f	ormat	

4. Specify the connection SIP address to be included in the To header on INVITE messages sent to the Unified Messaging server and click **OK**.

Note:

Typically, this address is a pilot number, the IP address, or FQDN of the Unified Messaging server. The port is 5060 for TCP or User Datagram Protocol (UDP), or 5061 for Transport Layer Security (TLS). This address is also used for connection if a proxy is not used.

For Exchange Server 2010, the **User Portion** must match the Pilot identifier for a Unified Messaging Hunt Group associated to the Unified Messaging IP Gateway created for the Customer Interaction Center server.

5. On the **Addresses** page in the **Configuration** dialog box, in the **Contact Line** box, click the SIP line created to handle communication between the Customer Interaction Center server and the Unified Messaging server.

For more information, see <u>Creating a line</u>.

Disabling delayed media

Some SIP devices do not support delayed media, which may cause calls initiated from a Customer Interaction Center client to fail to connect, immediately disconnect or have one-way audio. Therefore, you need to disable delayed media for INVITE and re-INVITE requests.

For information about accessing the Configuration dialog box, see Creating a station.

To disable delayed media:

1. In the navigation box in the **Configuration** dialog box, click **Session**.

Addresses	Like Clobal STP Station Service Settings	-
Audio Transport	i ose diobal sur station session settings	
Session	Use SIP Session Timer	
Authentication Region	SIP Session Timeout: 60 seconds	_
	SIP Register Interval: 1 Days	- I
	Disconnect on Broken RTP	
	Media Timing: Normal	
	Media reINVITE Timing: Normal	
	Terminate Analysis on Connect	
	Disable Media Server Passthru	
	Connection Call Warm Down Time: 5 seconds	
		-

2. In the Media Timing and Media reINVITE Timing boxes, select Normal.

Adding a custom attribute

By default, if a user authenticates with the Customer Interaction Center server, the user will be placed into the Interactive Voice Response (IVR) on the Customer Interaction Center server and will be required to dial a subscriber number to enter the IVR on the Unified Messaging server. You can change the default behavior of the Unified Messaging station by adding a custom attribute.

For information about accessing the Configuration dialog box, see Creating a station.

To add a custom attribute:

1. In the right pane of the **Stations** container, double-click the Unified Messaging station.

The Station Configuration dialog box is displayed.

2. Click the **Custom Attributes** tab.

tion Configuration - Unified Me	ion Options Custom Attributes Hist	cory
Name	Value	A <u>d</u> d
		<u>E</u> dit
		Delete
Path for Custom Attributes: \De	vLab\Production\CLAY\Workstations\l	Manage Attributes Jnified Messaging Station

3. Click Add.

The Add Custom Attribute dialog box is displayed.

Add Custo	om Attribute	×
<u>N</u> ame:	CUSTOM::	•
⊻alue:		
	ОК	Cancel

- 4. In the Custom box, click Divert for TUI.
- 5. In the **Value** box, type **1** and click **OK**.

Configuring Users

Configure Users to use a third-party Unified Messaging provider in the Interaction Administrator User container.

To configure a user:

- In the right pane of the Users container, double-click the name of the user to configure. The User Configuration dialog box is displayed.
- 2. Click the **Options** tab.

User Configuration -	Barbara.Martin				? ×
ACD Configuration Interaction Messa	MWI Licensing ge Store Quotas) C Personal Info Options	lient Configuration Workgroups Security	Roles Custom Attribute	etic Spellings Password Policies s History
Timeout for incomi Auto-Answer Non Fax Capability	ng interactions: -ACD Interactions		econds 💌	🔽 Use defau	JİC
Non-ACD Interact Alerting: Disconnected:	ion Actions	• •	Unified Messaging - Destination:	UM exchange Advanced Optic	▼ Ins
Calls Chats E-mails Generic Objects	Park	ed Call Setting: eout: ension:	s minute	S 🔽 Use defau	ılt
Confir	m auto-save		0	K Cance	

- In the Destination box, click the Unified Messaging station you created to handle communications routed between the Customer Interaction Center server and the Unified Messaging server.
 For more information, see <u>Creating a station</u>.
- 4. Click Advanced Options to display the Advanced Options dialog box.

Advanced Options				×
Unified Messaging User E	xtension	:		
Use IC Extension				
O Other:			l.	
	· ·			
ОК	1 [Cancel	1	
			_	

5. Specify the value to include in the From header on INVITE messages when accessing a user's mailbox or the Diversion header on INVITE messages when calls are transferred to the Unified Messaging server.

For Exchange Server 2010, provide the following:

For	Configure
Use IC Extension	Select this option when the user's Customer Interaction Center extension matches the user's Unified Messaging extension.
Other	 Select this option and provide one of the following: The user's Unified Messaging extension The E.164 address entered for the user, such as +13178723000 The SIP address for the user, prefixed with sip:.The address defaults to the user's e-mail address, such as sip:john.smith@mycompany.com.

Providing for failover and load balancing

There are two options for providing for failover and automatically balancing the load when multiple Unified Messaging servers are available. The first option is to set up a proxy for each of the Unified Messaging servers. The second option is to set up a single proxy that's used by all the Unified Messaging servers. For more information, see the Interaction Administrator online Help.

To set up multiple proxies:

- 1. In the right pane of the Lines container, double-click the line to configure.
- The Line Configuration dialog box is displayed.
- 2. In the navigation box on the SIP Line Configuration page, click Proxy.

Line Configuration	n	? ×
SIP Line Configura	ation Call Putback Custom Attributes History	
Line Audio Transport Session Authentication Proxy Registrar Headers Access Region Recorder	Prioritized list of Proxy IP addresses:	
	Add Edit Delete Up Down OK Cancel H	elp

3. Click Add.

The SIP IP Address dialog box is displayed.

SIP IP Address	5		×
IP Address:			
Port Number:	5060		
		ОК	Cancel

- 4. In the IP Address and Port Number boxes, specify the SIP IP Address and 5060.
- 5. Click OK.
- 6. Add a proxy for each Unified Messaging server.

ne Configuration SIP Line Configurat	- <stations-tcp></stations-tcp>	ustom Attributes History	1	? ×
Audio Transport Session Authentication Proxy Registrar Headers Access Region Recorder	IP Address 10.8.3.200 10.7.3.200 10.4.3.205	Delete Up Do	DNS SRV	
≪ 🔊 🗸 ⊆or	nfirm auto-save	c	K Cancel	Apply

- 7. On the SIP Line Configuration page, use the Up and Down buttons to arrange the proxies in the order in which they are to be used.
- 8. In the right pane of the **Stations** container, double-click the Unified Messaging station.

tation Configurati	on - Unified Messaging Statio	n	? ×
Configuration Acc	ess Control Station Options Cu	stom Attributes History	
Station Extension	: 111	✓ Active	
Addresses Audio Transport	Identification SIP Address:	111	<u>E</u> dit
Session Authentication Region	Connection SIP Address:	sip:111@111:5060	Edit
	Contact Line:	<use global="" setting="" sip="" station=""></use>] [
	Prefix for extension (optional):		
<u>≪</u>)≫	firm auto-save	OK Cancel	Apply

9. In the navigation box on the **Configuration** page, click **Transport** and enable the **Use Proxy for Station Connections** check box.

tation Configurati	ion - Unified Messag	ing Station	? ×
Configuration Acc	ess Control Station O	ptions Custom Attributes History	
Station Extension	. 111	I Acti⊻e	
Addresses Audio Transport Session Authentication Region	 □ Use Global SIP S □ Use Proxy for St Audio Protocol: Security: SIP DSCP Value: 	tation Transport Settings ation Connections RTP Minimal	
ແ 🚿 🔽 <u>C</u> on	firm auto-save	OK Cancel	Apply

To set up a single proxy:

- 1. In the right pane of the **Lines** container, double-click the line to configure.
- The Line Configuration dialog box is displayed.
- 2. In the navigation box on the SIP Line Configuration page in the Line Configuration dialog box, click Proxy.

Configuration	1			?
P Line Configural	tion Call Putback Custor	m Attributes History		
Line	Prioritized list of Proxy IP a	addresses:		
Audio	IP Address	Port	DNS SRV	
Session Authentication Proxy Registrar Headers Access Region Recorder			Use tel: Scheme	
	Add Edit Del	ete Up Down		

3. Click Add.

The SIP IP Address dialog box is displayed.

SIP IP Address	5		×
IP Address:			
Port Number:	5060		
		OK	Cancel

4. In the IP Address and Port Number boxes, specify the SIP IP Address and 5060.

SIP Line Configuration	- <stations-tcp></stations-tcp>	ustom Attributes History		? ×
Line Audio Transport Session Authentication Proxy Registrar Headers Access Region Recorder	IP Address 10.4.3.205	Port 5061	DNS SRV	
≫ 🔽 ⊆or	firm auto-save	Delete Up Down	Cancel	3pply

- 5. In the right pane of the **Stations** container, double-click the Unified Messaging station. The **Station Configuration** dialog box is displayed.
- 6. In the navigation box on the **Configuration** page, click **Transport** and enable the **Use Proxy for Station Connections** check box.

itation Configurati	on - Unified Messag	ing Station	? ×
Configuration Acc	ess Control Station O	ptions Custom Attributes History	
Station Extension	: 111	I ✓ Acti <u>v</u> e	
Addresses Audio Transport Session Authentication Region	 □ Use Global SIP S □ Use Proxy for St Audio Protocol: Security: SIP DSCP Value: 	tation Transport Settings ation Connections RTP Minimal	
<u>≪</u> ≫ ⊽ <u>C</u> onl	firm auto-save	OK Cancel	Apply

Encrypting SIP signaling and RTP streams

You can encrypt the SIP signaling stream and the RTP stream using TLS.

To enable encryption on the SIP signaling stream:

- In the right pane of the Lines container, double-click the line to configure. The Line Configuration dialog box is displayed.
- 2. In the navigation box on the SIP Line Configuration page, click Transport.
- 3. In the **Transport Protocol** box, click **TLS**.

P Line Conrigura	ition Call Putback Custo	m Attributes Histor	У	
Line Audio	Transport Protocol:	TLS 💌	_	^
Transport Session	Audio Protocol:	SRTP 💌	Security: End	d-to-Edge 💌
Authentication Proxy Registrar	Address to use:	Local Area Connec	tion	-
Headers			achine bus Network A	oapter
Access Region	Receive Port:	5061	Connect Timer (ms)	2000
Recorder TLS Security	Maximum Packet Retry:	4	T1 Timer (ms):	500
	Maximum Invite Retry:	3	T2 Timer (ms):	1000
	Reinvite Delay (ms):	750		
	Retryable Reason Codes:	480, 500-599		•

To enable encryption on the RTP stream:

- In the right pane of the Lines container, double-click the line to configure. The Line Configuration dialog box is displayed.
- 2. In the navigation box on the SIP Line Configuration page, click Transport.
- 3. In the Audio Protocol box, click SRTP.

For more information on these settings, see <u>Creating a line</u>, <u>Creating a station</u>, and *Interaction Administrator Help*.

Digital certificates

You can authenticate SIP/TLS connections to and from the Customer Interaction Center server and the Unified Messaging server using SIP and TLS line certificates. To support TLS, certificates must be configured on both the Customer Interaction Center server and the Unified Messaging server. The Customer Interaction Center server comes preconfigured with a Default Line Authority Certificate that can be used to sign third-party certificates. If a Unified Messaging server uses an existing certificate that has already been signed, you must import the Certificate Authority (CA) certificate on the Customer Interaction Center server. Use the TLS Security option on the SIP line to import the certificate. For more information, see Creating a line in this document.

The Unified Messaging server certificate is used for outbound connections, or when mutual authentication is required. Install the Default Line Authority Certificate as a trusted CA on the Unified Messaging server for inbound connections, or when mutual authentication is required. The Cipher Suites enabled on the Customer Interaction Center server must include at least one cipher suite supported by the Unified Messaging server.

SIP/TLS Line Certificates Configuration	? ×
Line Certificates Authority Certificates Port-To-Certificate Mappings Third Party Certificate Signing	
Here you can have your certificate signed by the signing authority certificate so that it can be properly authenticated when used on a SIP/TLS line.	
Authority Certificate: <pre></pre>	•
Certificate To Sign:	
	<u> </u>
	<u> </u>
Sign	
Clos	se
	~

Advanced setup

It is possible to use various combinations of multiple SIP lines and multiple Unified Messaging stations to either connect to multiple Unified Messaging servers or support both encrypted and unencrypted connections simultaneously.

For more information about configuration, see Enable Exchange Server 2010 MWI integration with Interaction SIP Proxy.

Exchange Server 2010

A single Unified Messaging IP Gateway can have multiple Unified Messaging Hunt Groups, each with a unique Pilot Identifier. This allows multiple Unified Messaging stations to be created, each with different settings. For more information, see <u>Creating a station</u>.

Enable Exchange Server 2010 MWI integration with Interaction SIP Proxy

Customer Interaction Center, when configured as a Unified Messaging (UM) IP gateway on the Exchange Server, does not natively support delivery of Message Waiting Indicator (MWI) Notify messages to endpoints when a user receives or deletes a message from his or her mailbox. However, you can support MWI messages in this environment if you install Interaction SIP Proxy and do some configuration tasks.

You can configure Interaction SIP Proxy to route MWI Notify messages to SIP endpoints, such as IP telephones, and to route voice mail call setup messages between Exchange Server 2010 and the Customer Interaction Center server.

For more information, see the following:

- Configuration limitations
- Install and connect Interaction SIP Proxy
- <u>Configure Interaction SIP Proxy</u>
- <u>Configure registration group in Interaction Administrator</u>
- Create a Unified Messaging station for the Exchange Server Pilot in Interaction Administrator
- Configure users in Interaction Administrator
- <u>Configure Exchange Server 2010 UM IP gateways and UM hunt group</u>
- Configure user mailboxes in Exchange Server 2010 UM

Configuration limitations

Using Interaction SIP Proxy to route MWI Notify messages from Exchange Server 2010 has the following limitations:

- Interaction SIP Proxy will be the destination for all Unified Messaging communications from the Customer Interaction Center server.
- On each Exchange Server 2010 machine, you must create Unified Message IP gateways for Interaction SIP Proxy and each Customer Interaction Center server, including fallback servers in a switchover configuration.
- The Pilot Identifier/Hunt Group that Customer Interaction Center will use will be associated to the Unified Messaging IP Gateway for the Interaction SIP Proxy system.
- Interaction SIP Proxy will have a From based route configured to send Customer Interaction Center traffic to the Exchange server.
- The Unified Messaging extension for the user will match the Customer Interaction Center user extension.
- Polycom phones need to register to the Customer Interaction Center servers as well as the Interaction SIP Proxy system.
- You must set a default station for each user defined in Customer Interaction Center to which you want to send MWI messages.
- The voice mail button on the telephone will not retrieve the voice mail messages for a user.
- If a user logs on or off a station, the MWI indicator will not change.
- The MWI indicator on the phone is enabled when the e-mail message arrives in the mailbox of the user.

Install and connect Interaction SIP Proxy

- 1. Through Genesys or your Genesys partner, purchase Interaction SIP Proxy.
- 2. Install Interaction SIP Proxy.

For information on installation requirements and procedures, see "Interaction SIP Proxy installation or upgrade" in *Interaction SIP Proxy Technical Reference*.

3. In the Interaction SIP Proxy web interface, configure Interaction SIP Proxy to connect to the Customer Interaction Center server by selecting **System > IC Integration**.

For more information, see "Connect Interaction SIP Proxy to the Customer Interaction Center server" in *Interaction SIP Proxy Technical Reference*.

4. Using Interaction Administrator, trust the connection from the Interaction SIP Proxy system.

For more information, see "Trust the Interaction SIP Proxy connection and assign a location" in Interaction SIP Proxy Technical Reference.

Configure Interaction SIP Proxy

- 1. Open a web browser and navigate to the URL address of the Interaction SIP Proxy address server.
- 2. Enter the user name and password for Interaction SIP Proxy and select the OK button.
- 3. In the upper right corner, select the Config icon.
- 4. On the General tab, set the following controls as indicated:
 - Digest Authentication No
 - Record Route Yes
 - Rewrite Request URI Yes
 - Registration has Priority Yes
- 5. Select the Server Plan tab.

The Config-Server Plan page is displayed.

- 6. On the right side of the page, select the **insert** hyperlink.
 - A set of controls is displayed.
- 7. In the box at the top of the set, enter a name for this server plan. Use a name, such as "Exchange UM MWI" or something similar, so that you can identify this entry easily in the future.
- 8. In the **Destinations** box, enter the address of the Exchange Server 2010 system. You can specify an IP address, host name, or fully-gualified domain name.
- 9. Change other settings, such as **Protocol** and **Timeout**, as necessary.

For more information on these controls, see *Interaction SIP Proxy Help* by selecting the **Help** icon in the upper right corner of the page.

- 10. Select the **Apply** button.
 - The entry is saved and displayed.
- 11. Select the **Routing** tab.

The Config-Routing page is displayed.

- 12. On the right side of the page, select the **insert** hyperlink.
- A set of controls is displayed.

13. In the box to the right of the check box, enter a name for this route.

Note:

Include "IC UM MWI" or something similar in this name to ensure you can identify it easily in the future.

14. In the From Address box, enter the following text:

(.*)@server_address

server_address is the address of your Customer Interaction Center server. You can specify an IP address, host name, or fullyqualified domain name.

15. In the To Address box, enter the following text:

.*

- 16. In the **Routes to** box, enter **\$1**.
- 17. In the list box to the right of the **Routes to** box, select the server plan that you created earlier in this procedure.

- 18. In the **Description** box, you can, optionally, enter text to further identify this route.
- 19. Select the **Apply** button.
 - The route entry is saved and displayed.
- 20. Ensure that the check box for the route you created is enabled.

Note:

If you use a switchover pair of Customer Interaction Center servers, you must repeat steps 11 through 20 for the second server.

21. Select the Authentication tab.

The Config-Authentication page is displayed.

22. On the right side of the page, select the **insert** hyperlink.

The following set of controls is displayed.

23. In the **Registration Id** box, enter a unique, identifiable string for a managed IP telephone that you want to receive MWI messages.

For more information on defining and configuring managed IP telephones in Interaction Administrator, see "Home > Interaction Administrator > IC Server > Managed IP Phones" in *Interaction Administrator Help*.

24. In the Username box, enter the user name for the specified station.

Note:

You set the user name and password for a station through the Stations container in Interaction Administrator.

- 25. In the **Password** box, enter the password for the specified station.
- 26. In the Aliases box, enter the extension that is assigned to the user of this station.

Note:

You set the extension number for a user through the **Users** container in Interaction Administrator.

27. On the right side of the page, select the **Apply** button.

The authentication entry is saved and displayed.

Configure registration group in Interaction Administrator

To receive MWI messages, each managed IP telephone must use a Customer Interaction Center registration group with the appropriate configuration. To create the registration group, do the following steps:

- 1. Log on to Interaction Administrator.
- 2. In the left pane, expand the container for your Customer Interaction Center server.

🏶 Interaction Administrator - [Examp	le]
📄 Eile Edit View Context Help	
🕞 🔉 🔊 🗂 🕲 🎾 🔒 📗	
🖃 🏇 Collective	Name
🞊 Home Site	<default group="" registration=""></default>
🔿 Peer Sites	<default gro<="" registration="" secure="" td=""></default>
🖃 🛒 EXMP -	
🛸 Lines	
🖈 Line Groups	
🕀 🛄 Stations	
🕀 🜆 Managed IP Phones	
👷 Registration Groups	
TP Bridges	
👲 Audio Sources	
🎆 Server Parameters	
🗐 Structured Parameters	
🕀 🌍 Regionalization	
📄 Licenses Allocation	
🕀 🏥 People	

- 3. Select the **Registration Groups** container.
- 4. In the right pane, right-click an open area and select **New** from the resulting shortcut menu. The **New Registration Group** dialog box is displayed.

New Registrati	ion Group	
<u>N</u> ame:	1	
<u>Type</u> :	Regular	•
	ОК	Cancel

- 5. In the Name box, enter a unique, identifiable name for this IP telephone registration group.
- 6. In the **Type** list box, select the **Regular** entry.
- 7. Select the **OK** button.

The Registration Group Configuration dialog box is displayed.

Registration Gro	up Configuration - MWI Registration Group	? ×
Configuration C	Options	
Manage Add add	d IP phones will send communications through the registrations in th litional registrations for redundancy.	ne specified order.
Name: MWI	Registration Group	_
Type: Regula	ır	
Registrations -		
Туре	Details	A <u>d</u> d
		<u>E</u> dit
		<u>R</u> emove
		Move Up
		Move Down
	C	DK Cancel

8. Select the Add button.

The Add Registration dialog box is displayed.

Registration				?
Obtain registration s	settings auto	omatically fro	m this line: –	
Line:	<stations-< th=""><th>UDP></th><th></th><th>•</th></stations-<>	UDP>		•
<u>U</u> se the following re	gistration se	ttings:		
Address:				
Port:	5060			
Transport Protocol:	🖸 UDP	O TOP	O TLS	
Alternate Address:	SIP Proxy	1		
Transport Protocol:	🖲 UDP	C TCP	C TLS	
Obtain registration s	settings auto	omatically usi	ing <u>D</u> NS SRV:	
Obtain registration s	settings auto	omatically usi	ing <u>D</u> NS SRV:	
Obtain registration s Domain: Transport Protocol:	Settings auto	O TCP	O TLS	

- 9. Select the Obtain registration settings automatically using DNS SRV option.
- In the Domain box, enter domain in which the Customer Interaction Center server resides.
 For example, if the fully qualified domain name of the Customer Interaction Center server is CICserver1.mycompany.com, enter mycompany.com in the Domain box.
- 11. In the Transport Protocol area, select the TCP option.
- 12. Select the **OK** button. The DNS SRV entry is saved and displayed in the **Registration Group Configuration** dialog box.
- 13. In the **Registration Group Configuration** dialog box, select the **Add** button.
- The Add Registration dialog box is displayed.
- 14. Select the Obtain registration settings automatically from this proxy option.
- 15. In the SIP Proxy list box, select the Interaction SIP Proxy server that you installed and configured previously in this procedure.
- 16. In the **Alternate Address** box, enter an address that the managed IP telephone can use to send SIP messages to the Interaction SIP Proxy server.

Important!

Specify the information in the **Alternate Address** box only if the IP telephones exist in locations other than the one that contains this Interaction SIP Proxy server.

- 17. In the Transport Protocol area, select the TCPoption.
- 18. Select the **OK** button.

The Interaction SIP Proxy entry is saved and displayed in the Registration Group Configuration dialog box.

egistration Grou	o Configuration - MWI Registration Group	p <u>?</u>	×
Configuration Op	tions		
Managed IP phones will send communications through the registrations in the specified order. Add additional registrations for redundancy.			
Name: MWI Re	gistration Group		
Type: Regular			
Registrations —	Details		
DNS SRV	Details mycompany.com (TCP)	A <u>a</u> a	
SIP Proxy	SIP Proxy 1 [172.17.110.86] (TCP)	<u>E</u> dit	
		<u>R</u> emove	
		Move <u>U</u> p	
		Move Down	
« 🔊 🔽 🗠	ifirm auto-save	OK Cancel Apply	
Important!			

Create a Unified Messaging station for the Exchange Server Pilot in Interaction Administrator

Important!	
This procedure requires that you have created a Pilot Identifier number on the Exchange Server 2010 UM.	

- 1. Log on to Interaction Administrator.
- 2. In the left pane, expand the configuration for your Customer Interaction Center server.

Ensure that the DNS SRV entry is listed first in the Registrations box.

- 3. Select the **Stations** container.
- 4. In the right pane, right-click an open area and select **New** from the resulting shortcut menu.

The Entry Name dialog box is displayed.

Entry Name		? ×
Enter Station Name		
1.		
	ОК	Cancel

- 5. In the Enter Station Name box, enter EXCHANGE.
- 6. Select the **OK** button.
 - The **Station Type** dialog box is displayed.

Station Type			×
Station Type: Remote Station			
Stand-alone Fax			
Unified Messagir	ne		
Workstation			
		Unified Messaging Provider	:
Unified Messagi	ng	Exchange	-
Select the station template that will be used create a new station from scratch.	for the new station. T	'his can be left blank, if you wis	h to
Station Template:	Ŧ		
	< Back Next :	> Finish C	Cancel

- 7. In the Station Type box, select Unified Messaging.
- 8. In the Unified Messaging Provider list box, select Exchange.
- Select the Next button.
 The Configuration dialog box is displayed.

Configuration			? ×
Station Extension	: Aut	to_	
Addresses Audio Transport Session	Identification SIP Address:	No SIP address is configured	<u>E</u> dit
Authentication Region	Connection SIP Address:	No SIP address is configured	E <u>d</u> it
	Contact Line:	<use global="" setting="" sip="" station=""></use>	
	Prefix for extension (optional):		
	< <u>B</u> ack	Next > Finish	Cancel

- 10. In the **Station Extension** box, enter the number for the Exchange Pilot Identifier.
- Select the Edit button to the right of the Identification SIP Address box. The Identification SIP Address dialog box is displayed.

Identification SIP Address	×
Use a predefined format (sip:user@host:port)	
Use User Portion Only	
User Portion: Host: Port:	_
5060	
Address:	
sip:@:5060	
O Use an alternate format	
Address:	_
OK Cancel	

- 12. In the **User Portion** box, enter the number of the Exchange Pilot Identifier.
- 13. In the Host box, enter the address of the Interaction SIP Proxy server that you configured earlier in this procedure.
- 14. Select the **OK** button.

The address is displayed in the **Configuration** dialog box.

15. In the **Configuration** dialog box, select the **Edit** button to the right of the **Connection SIP Address** box. The **Connection SIP Address** dialog box is displayed.

Connection SIP Address	×
• Use a predefined format (sip:user@host:port)	
User Portion: Host: Port:	
Address: sip:@:	_
C Use an alternate format	
Address;	
OK Cancel	

- 16. In the User Portion box, enter the number of the Exchange Pilot Identifier.
- 17. In the Host box, enter the address of the Interaction SIP Proxy server that you configured earlier in this procedure.
- 18. In the **Port** box, enter **5060**.
- 19. Select the **OK** button.

The address is displayed in the **Configuration** dialog box.

20. In the **Contact Line** list box of the **Configuration** dialog box, select the Customer Interaction Center SIP line that you want to use for this station.

Important!

The SIP line that you select for Contact Line must be configured to use the TCP protocol and port 5060.

21. In the **Configuration** dialog box, select the **Next** button. The **Access Control** dialog box is displayed.

Available Blocked Emergency	Currently Selected
Information Intercom International	Add >
Local Operator Long Distance SIP	
TEL Toll Free Unknown	

- 22. In the **Available** box, select **SIP** and then select the **Add** button.
- 23. Select the **Finish** button.

The station for Exchange Unified Messaging is saved and displayed in Interaction Administrator.

Configure users in Interaction Administrator

Note:

You must do this procedure for each user who you want to receive MWI messages from Exchange Server 2010 UM.

- 1. Log on to Interaction Administrator.
- 2. In the left pane, expand the container for your Customer Interaction Center server.
- 3. Select the **Users** container.

🕞 Interaction Administrator - [Example]			
Eile Edit View Context Help			
🕞 👂 🖹 🗈 🛍 🚳 🎾 🔒			
🖃 🏇 Collective	User Name		
🔊 Home Site	admin_user		
🄝 Peer Sites	Operator		
🕀 🛒 EXMP -			
Licenses Allocation			
🖃 🎒 People			
🦥 Default User			
Roles			
🕹 Users			
Sea Workgroups			
🛨 🦿 Password Policies			
Schedules			

4. Double-click a user entry that you want to configure to receive MWI messages from Exchange Server 2010 UM. The **User Configuration** dialog box is displayed.

User Configuration - Ke	rry Smith				? ×
Client Configuration	Phonetic Spellings	Options	Security	Custom Attributes	History
Configuration Licens	ing Personal Info	Workgroups	Roles Pa	assword Policies AC	D MWI
Extension:	11	Mailbox Use	r: Ke	erry Smith	
Password:	•••••	Preferred La	anguage: <a>S	ystem Default>	•
Confirm Password:		Default Wor	kstation: Cla	ау	•
IC Privacy Name: Ke	rry	NT Domain l	Jser:		
Outbound ANI: 55	5-1234			Exclude From Director	y
Location:	efault Location >		•		
Time Zone:	TC-05:00) Eastern Time	(US & Canada)		Use Location Time Zon	e
Microsoft Lync Setting	gs	Looku	Settings		_
User Address.		LOOKU			_
Line URI:		Use	r Type: En	terprise Voice	I
Enable Presence Synchronization					
Current Status:	vailable	Home Site:	11	11 (DevLab)]
		Current Site	: <	Not Set>]
Confirm a	auto-save		ОК	Cancel	Apply

- 5. In the Extension box, enter the UM Extension configured in Exchange Server 2010 UM.
- 6. In the **Default Workstation** list box, select the station that will receive MWI messages for this user.
- 7. Select the **Options** tab.

The **Options** tab is displayed.

User Configuration - Kerry Smil	th				? ×
Configuration Licensing Pe Client Configuration Phone	ersonal Info etic Spellings	Workgroups Options	Roles Pa Security	assword Policies ACD Custom Attributes	MWI History
Incoming Faxes Fax capability Web Client fax format:			y	☑ Use default	
Unified Messaging			▼ Adv	anced options	-
Non-ACD Interactions Auto-Answer Non-ACD I Alerting Action:	interactions		•		_
Incoming Interactions	<u></u>	Seconds	-	🔽 Use default	_
Confirm auto-save			ОК	Cancel	Apply

- 8. In the **Destination** list box in the **Unified Messaging** area, select the **EXCHANGE** registration group that you created previously in this procedure.
- 9. Select the **OK** button.

The changes to this user entry are saved.

Configure Exchange Server 2010 UM IP gateways and UM hunt group

- 1. Open the Exchange Management Console window.
- 2. In the left pane, expand the Organization Configuration container.
- 3. In the left pane, select the Unified Messaging item.

The Unified Messaging pane is displayed on the right side.

- 4. Select the UM IP Gateways tab.
- 5. Using the Add Expression button, add the following servers:
- Primary Customer Interaction Center server
- Secondary Customer Interaction Center server, if using a switchover pair
- The Interaction SIP Proxy server that you configured earlier in this procedure

Important!

Ensure that you set the **Pilot Identifier** for the default UM hunt group of the Interaction SIP Proxy server UM IP Gateway to the value you specified in the <u>Create a Unified Messaging station for the Exchange Server Pilot in Interaction Administrator</u>.

Configure user mailboxes in Exchange Server 2010 UM

In Microsoft Exchange Server 2010 UM, configure each user mailbox to have the UM extension that you specified for that user in Interaction Administrator.

Kerry Smith UM Properties	×
Kerry Smith UM Properties General User Information Addition Mail Flow Settings Mail Mail Flow Settings Mail Properties Enable D Feature State Outlook Web App Enable Exchange ActiveSync Enable D Unified Messaging Enable Enable MAPI Enable Enable MAPI Enable Discription Description To enable or disable Unified Messaging''. Enable Unified Messaging is a premium for the action pane, select "Enable Unified Messaging''.	Image: Section
Unified Messaging is a premium f Client Access License (CAL) to e	ie ature and req ✓ Allow the user to receive faxes nable it for the ✓ Allow diverted calls without a caller ID to leave a message ✓ Allow users to configure call answering rules Personal operator extension: 11111 Cancel OK Cancel

Change Log

The following table lists the changes to the Unified Messaging Technical Reference since its initial release.

Date	Changes
01-October-2011	Initial Release
23-July-2013	 Updated formatting Added content for supporting MWI messages from Exchange Server 2010 UM with Interaction SIP Proxy
05-September-2014	Updated documentation to reflect changes required in the transition from version 4.0 SU# to CIC 2015 R1, such as updates to product version numbers, system requirements, installation procedures, references to Interactive Intelligence Product Information site URLs, and copyright and trademark information.
13-October-2015	 Updated cover page to reflect corporate branding Updated "Copyright and trademark information" page Minor edits
08-August-2017	 Removed Exchange Server 2010 TLS cipher suites section from Digital certificates topic. Updated copyright and trademark page.
10-May-2018	Rebranded from Interactive Intelligence to Genesys
14-June-2019	Reorganized the content only, which included combining some topics and deleting others that just had an introductory sentence such as, "In this section".