

**MiCollab Advanced Messaging
Verizon Hosted IP Centrex (HIPC)
SIP Trunk
Integration Technical Note**

For version 6.1 and above

Notice

The information contained in this document is believed to be accurate in all respects but is not warranted by Mitel Networks™ Corporation (MITEL®). Mitel makes no warranty of any kind with regards to this material, including, but not limited to, the implied warranties of merchantability and fitness for a particular purpose. The information is subject to change without notice and should not be construed in any way as a commitment by Mitel or any of its affiliates or subsidiaries. Mitel and its affiliates and subsidiaries assume no responsibility for any errors or omissions in this document. Revisions of this document or new editions of it may be issued to incorporate such changes.

No part of this document can be reproduced or transmitted in any form or by any means - electronic or mechanical - for any purpose without written permission from Mitel Networks Corporation.

Trademarks

The trademarks, service marks, logos and graphics (collectively "Trademarks") appearing on Mitel's Internet sites or in its publications are registered and unregistered trademarks of Mitel Networks Corporation (MNC) or its subsidiaries (collectively "Mitel") or others. Use of the Trademarks is prohibited without the express consent from Mitel. Please contact our legal department at legal@mitel.com for additional information. For a list of the worldwide Mitel Networks Corporation registered trademarks, please refer to the website: <http://www.mitel.com/trademarks>.

© Copyright 2016, Mitel Networks Corporation

All rights reserved

Contents

Preface	4
References	5
Documentation	5
Documentation Updates	5
Help	6
Document Conventions	6
Features Supported by this Integration	7
Critical Application Considerations	9
Installation Requirements	11
Telephone System Requirements for Verizon HIPC	11
MiCollab AM Requirements	11
Programming the Telephone System	12
Configuring MiCollab AM	13
Configuring MiCollab AM for the Integration During Initial Installation	13
Completing the Integration	16
Configuring Existing MiCollab AM for the Integration	16
Configuring MiCollab AM for SIP Failover	19
Changing the Network Binding Order on the MiCollab AM Platform	21
Windows Server 2008 R2 with Service Pack 1	21
Windows Server 2012 R2	22
Configuring Quality of Service (QoS)	23

Preface

This Integration Technical Note (ITN) is written for dealers who are experienced with MiCollab Advanced Messaging (MiCollab AM) and are familiar with its procedures and terminology. This document also assumes that you are familiar with the features and programming of the Verizon Hosted IP Centrex (HIPC) telephone system, referred to in this document as Verizon HIPC.

This document describes how to integrate MiCollab AM with a Verizon HIPC telephone system, using the Session Initiation Protocol (SIP) integration. This integration operates exclusively over a TCP/IP-based network; it uses no analog or digital voice telephony ports, but passes voice communication and signaling information over the network.

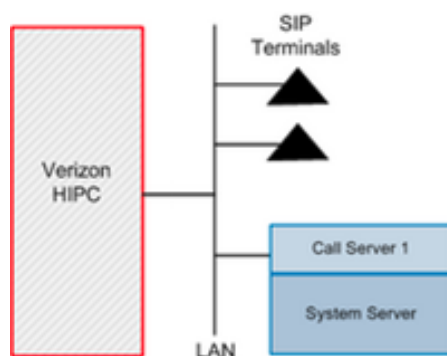


Figure 1. Verizon HIPC Integration

The Verizon HIPC SIP Trunk integration consists of the telephone system and MiCollab AM. Verizon configures the SIP trunks as virtual trunks and terminates them on the dedicated facilities at the site. MiCollab AM uses the SIP trunks to integrate with the switch through static SIP endpoints. The Verizon HIPC routes all calls directed to MiCollab AM to these SIP endpoints. Similarly, the Verizon HIPC routes all of the outgoing calls and MWI requests from MiCollab AM to their proper destinations. MiCollab AM provides the hunting.

The Verizon HIPC can direct calls to multiple MiCollab AM Call Servers through the Mitel SIP Routing Manager Service. The SIP Routing Manager Service:

- Changes the way in which SIP calls are directed; it does not change the destination of either inbound or outbound calls.
- Acts as an intermediary, it redirects SIP call traffic to the appropriate destination.
- Allows the routing of SIP Trunk calls to multiple MiCollab AM Call Servers

For more information on the SIP Routing Manager Service, refer to the Configuring MiCollab AM section.

This ITN documents the procedure for setting up the integration. The process consists of programming the telephone system and configuring MiCollab AM. If the SIP Routing Manager is deployed for this integration, it must be installed and configured prior to starting MiCollab AM. Critical application considerations are also documented.

References

A catalog of technical documentation is included on the MiCollab AM Installation Media. If you are installing any advanced applications, such as Networking and Fax Server applications, you should refer to the appropriate technical documentation for application and installation information.

Documentation

The technical documentation is produced in the PDF format and requires the PDF reader to view it. The documentation set for this MiCollab AM includes the following documents and resources:

- **Developer Resources.** Contains programming guides and API references for developers for integrating the server clients and web applications with MiCollab AM.
- **Integration Technical Notes (ITN).** Contains a set of guides that describe the integration methods and instructions for a variety of phone systems to work with MiCollab AM. The ITNs are generally used by resellers or administrators who are experienced with MiCollab AM and familiar with the integration procedures and terminology.
- **Quick Reference Card (QRC).** Contains shortcuts and quick instructions telling subscribers how to access and use the messaging system.
- **Server Documentation.** Available as a PDF only. Contains administrative guides for administrators about installing, configuring, and administering the messaging system, and user guides for subscribers about accessing the messaging system and checking and sending messages.
- **Spare Parts Documentation.** Contains a set of guides that describe the instructions for installing and configuring hardware parts to work with MiCollab AM. These documents are written for Mitel certified MiCollab AM technicians who are experienced with MiCollab AM and familiar with the procedures and terminology.
- **Software Release Notice (SRN).** This notice introduces the new features, capabilities, and hardware/software requirements for the corresponding MiCollab AM version.

For more detailed documents, refer to the following list of references:

Table 1. References

Document Type	Document Title
Server Documentation	System Administration Guide
Server Documentation	System Installation Guide
Server Documentation	SIP Routing Manager Installation Guide

Documentation Updates

Documentation updates may be available from the following sources:

- Mitel certified technicians can view or download the latest/updated documents and program files from our partner web site: connect.mitel.com/connect

Help

The primary source of information about MiCollab AM is the online help available within any of its administrative utilities. You can access **Help** as follows:

- Click the **Help** button in the dialog box or window in which you are working
- Press the **F1** key at any time.

Document Conventions

The following conventions are used in this document:

- **Key Names.** Names of keys on the keyboard are shown in a box.

Example: **Enter**

When two keys must be pressed simultaneously, they are joined by a + sign.

Example: **Alt** + **Tab**

- **Reference to Document.** *Italics* fonts can also signify the titles of other documents.

Example: Refer to *System Installation Guide*.

- **UI Element Names.** Names of UI elements such as dialog windows, screens, menu items, tabs, buttons, icons, etc. are shown in bold.

Example: On the **Startup** screen, click the **Start** icon.

- **User Input.** Information required to be typed is shown in italics.

Example: Type the password *voicemail*.

- **Warning, Caution, Important, and Notes.** Text for the contents that require attention are shown as follows:

WARNING A warning paragraph advises you of circumstances that can result in the loss of data, harm to the system server platform, or personal harm.

CAUTION Failure to follow these recommendations can result in unauthorized access to the system and consequent loss of data.

IMPORTANT An important paragraph gives decision-making information or informs you of the order in which tasks need to be completed.

NOTE A note gives additional information, provides an explanation, or indicates an exception to the information in the preceding text.

Features Supported by this Integration

The following tables list the features supported using the Verizon HIPC SIP Trunk integration.

Table 2. Call forward to personal greeting support for these common call types

Divert to MiCollab AM on	Supported
No Answer	Yes
Busy	Yes
Forward All	Yes
Do Not Disturb	No

Table 3. Integration features supported for Verizon HIPC SIP Trunk

Feature	Supported	Notes
Automatic subscriber logon	Yes	
ANI/CLI	Yes	
Announce Busy greeting on forwarded calls	Yes	
Call screening	Yes	Note 1
Caller queuing	Yes	Note 2
DNIS	Yes	
End-to-end DTMF, attendant console	Yes	
End-to-end DTMF, proprietary telephones	Yes	
Fax Tone Detection	Yes	
Internal calling party ID for reply	Yes	
Live record, integrated	No	
Live reply to sender	Yes	
Message notification callouts	Yes	
MWI, set/clear	Yes	
MWI, inband/outband	Outband	

Networking; AMIS, analog	No	
Overflow from MiCollab AM to attendant	Yes	
Overflow to MiCollab AM from attendant	Yes	
PBX-provided disconnect signaling	Yes	
Revert to operator	Yes	
Transfers, blind	Yes	Note 3
Transfers, confirmed	Yes	Note 3
Transfers, fully supervised	Yes	Note 3
Transfers, monitored	Yes	Note 3
Trunk ID for call routing	No	
SIP Routing Manager	No	
Multiple Integrations	Yes	Note 4
Automatic subscriber logon	Yes	

NOTES

1. Only available when using supervised transfers.
2. Caller Queuing is specific to each local Call Server. Call Servers within the system are unaware of queued calls to the same subscriber on other Call Servers. For more information, refer to [Critical Application Considerations](#).
3. When transferring to TDM devices. For more information, refer to the information regarding transfers in [Critical Application Considerations](#).
4. See [Critical Application Considerations](#).

Critical Application Considerations

Known limitations or conditions within the telephone system and MiCollab AM that affect the integration performance are listed here. General recommendations are provided when ways to avoid these limitations exist.

- If you want to call individual lines directly, you must populate the line extensions on the Lines tab before starting MiCollab AM.
- You must configure the Incoming Hunt Mode in the Switch Section Options dialog box. This integration supports terminal, circular, reverse terminal and reverse circular hunt modes only. The default mode is Terminal.
- You must configure the Trunk Group Access Code in the Switch Section Options dialog box. This code cannot conflict with extensions. This must match with the DN prefix associated with this MiCollab AM Call Server.
- The SIP Domain Name in the Integration Options dialog box must match the domain name configured on the Verizon HIPC. This is a case sensitive value.
- The primary (public) network interface card (NIC) must be the first network connection in the network binding order. MiCollab AM binds and communicates to other servers and subscribers on this network connection. On a MiCollab AM server with two or more NICs, the NIC that supports this integration must not occupy first place in the operating system's binding order. For more information, refer to the [Changing the Network Binding Order on the MiCollab AM Platform](#) section later in this document.
- If a call is transferred out of MiCollab AM, and at least one end-point is a TDM based device, the inbound and outbound SIP trunks remain connected for the duration of the call. The MiCollab AM port returns to idle and can take other calls as long as additional SIP resources are available. Because additional SIP resources are required, a SIP trunk to MiCollab AM port ratio greater than 1:1 is recommended.
- MiCollab AM supports G.729a with support for annex b on the incoming audio stream only. MiCollab AM does not transmit annex b packets.
- When codec negotiation takes place between MiCollab AM and the PBX, MiCollab AM always offers the G.729 audio format as an option. You may configure G.729 as the preferred codec in MiCollab AM; however, the decision whether to use G.729 is always made by the PBX.
- The Call Queuing feature does not transcend the Call Server. Calls may be queued on multiple Call Servers for the same subscriber but Call Servers do not have knowledge of calls in the queue on other Call Servers within the system. Callers may be prompted with specific information about their place in the queue; however, the information pertains to the specific Call Server on which their call is queued.
- Discuss the network switch and firewall requirements with the site's IT administrator before you begin the integration with MiCollab AM to resolve any restrictions related to ports and IP addresses through the premise firewall, routers, and network.

- The MiCollab AM Integration Options parameter, "Validate Remote Hosts for Media" validates each incoming audio packet and accepts it only if it is sent from a valid endpoint. The parameter is disabled by default. Enabling this parameter causes MiCollab AM to reject RTP packets from invalid endpoints, rejects MWI packets that timeout after a specified number of times, and overcomes port lockups when callers hang up while MiCollab AM is performing a blind transfer. See the Mitel Technical Bulletin #42285 for more information on this parameter.

NOTE Enabling this parameter causes processing overhead, enable it only when it is necessary.

- MiCollab AM 6.1 supports up to 10 integration types (i.e. licensed integrations) in total per system. However, the following limitations apply to each Call Server:
 - Limited to 3 integration types per Call Server
 - The 3 integration types can be any mix of TDM and SIP (e.g. 1 TDM and 2 SIP)
 - Limited to 1 Mitel MiTAI or 1 Cisco UCM SCCP IP integration. Can be mixed with TDM, but not with SIP.
 - Connect up to 10 telephone systems total per Call Server (e.g. 2 Avaya Communication Manager systems using SIP + 5 Avaya IP Office systems using SIP + 3 Siemens HiPath 4000 systems using Station Set Emulation)

Installation Requirements

Review the following information before performing any of the procedures in this document. To install this integration successfully, you must meet the installation requirements for both the telephone system and MiCollab AM.

Telephone System Requirements for Verizon HIPC

- Verizon HIPC is not available in all areas. Please contact your Verizon Business Representative for feature availability and configuration information. You can learn more about HIPC from the Verizon website, <http://www.verizonbusiness.com>.

MiCollab AM Requirements

- MiCollab AM version 6.1
- Mitel software key diskette or feature file with the Verizon HIPC SIP Trunk integration enabled and one RADVISION® SIP and RTP license enabled for each port involved in the integration
- One 100 Mbps or 1000 Mbps (1 Gbps) network interface card

Programming the Telephone System

Please contact your Verizon Business Representative for feature availability and configuration information. Your business representative will help you determine the type of physical interface, network gateway, and feature set you require for your site. Verizon technical staff will work with your telecomm staff to implement your installation.

Configuring MiCollab AM

Once the telephone system is programmed, you must configure MiCollab AM for the integration. There are two ways you can configure MiCollab AM: (1) Configuring MiCollab AM for the telephone system integration when you are installing MiCollab AM for the first time, or (2) Configuring the existing MiCollab AM with the new telephone system integration.

Click the appropriate steps that your system requires from below and follow the steps:

- [Configuring MiCollab AM for the Integration During Initial Installation](#): Integrate the telephone system while you install MiCollab AM for the first time.
- [Configuring Existing MiCollab AM for the Integration](#): Integrate a new telephone system on your existing MiCollab AM system.

NOTE If you want to call individual lines directly, you must populate the line extension numbers on the Lines tab before starting MiCollab AM.

Configuring MiCollab AM for the Integration During Initial Installation

To configure MiCollab AM for the integration during the initial installation:

- 1 In the **Database Initialization Parameters** dialog box, configure the following options:
 - a In the **Mailbox Length** box, enter the mailbox length in digits.
 - b In the **First Extension** box, enter first extension number for the first line. You can also leave the **First Extension** box empty.
 - c From the **Manufacturer** dropdown list, select **Verizon**.
 - d From the **Model** dropdown list, select **HIPC**.
 - e From the **Integration Type** dropdown list, select **SIP Trunk**.
- 2 Click **Next**. The **Board Options** dialog box displays.
- 3 In the **Board Options** dialog box, configure the following options:
 - a From the **Manufacturer** dropdown list, select **RadVision**.
 - b From the **Model** dropdown list, select **SIP STACK**.
 - c In the **Name** field, the name for this board is automatically generated. Enter a new name if necessary.
 - d From the **Protocol** dropdown list, select **SIP IP RTP**.

- e In the **Number of Lines** field, enter the number of lines this board uses. The total number of lines is limited by the capacity of the board and the number of **Available Line Licenses**.

- 4 Click **OK**. The **Switch Options** dialog box displays.

NOTE The settings related to the telephone system in the **Switch Options** dialog box are filled in automatically when you select the correct telephone system during setup.

- 5 If necessary, make any changes to the default settings your site requires in the **Switch Options** dialog box.

NOTE If you need to customize settings on the **Switch Options** dialog box to meet requirements specific to your site, refer to the documentation accompanying the telephone system, the online help, and the guide, *System Installation Guide*.

- 6 Click **OK**. The **Integration Options** dialog box displays.

- 7 In the **Integration Options** dialog box, configure the following options:

- a In the **Local Integration Settings** section, select the **Required Parameter** view and configure the following options:

Table 4. Required Parameters Settings – Integration Options

Field	Value
SIP Server Address	If... Then...
	The SIP Routing Manager is not installed Enter the Verizon HIPC IP address.
	The SIP Routing Manager is installed Enter the SIP Routing Manager IP address.
SIP Server Port	If... Then...
	The SIP Routing Manager is not installed Enter the listen port configured on the Verizon HIPC.
	The SIP Routing Manager is installed Enter the listen port configured on the SIP Routing Manager.
Transport for outgoing SIP messages	Enter UDP , the default.
Use DNS Discovery Process	Select this box to use DNS discovery.
Local IP Address to bind on	Select the IP address of the NIC on the Call Server platform that connects to the Verizon HIPC.

SIP Local Connection Port	Enter the TCP port MiCollab AM listens for incoming SIP messages. The default port is 5060.
SIP Parser Qualifier String	<p>In cases of a single SIP integration on the call server, enter the local IP address to which the integration is bound. This field is used by MiCollab AM to match SIP packets to the appropriate SIP integration.</p> <p>In cases where there are multiple SIP integrations on the call server, use a string that is unique to each SIP integration.</p> <p>For example:</p> <p>The extension that will be used as the hunt number on the PBX followed by the "@" symbol and the IP of the call server, such as 5000@172.16.4.202.</p> <p>The hunt number must be unique across all IP integrations.</p> <p>The Fully Qualified Domain Name (FQDN) of the switch, such as pbx1.sipdomain.com.</p> <p>NOTE This setting must match a string in the SIP header that is unique to this particular integration.</p>

- b** Click **OK**. The **Switch Section Options** dialog box displays.
- 8** In the **Switch Section Options** dialog box, configure the following options:
 - a** In the **Local Switch Section Settings** section, select the **Required Parameters** view.
 - b** In the **Incoming Hunt Mode** field, select the mode for this integration.

NOTE This integration supports **terminal**, **circular**, **reverse terminal**, and **reverse circular** hunt modes only.

 - c** In the **Hunt Group Access Code** field, enter the Trunk Group Access Code. This number must match with the Directory Number that Verizon has specified as the trunk access number for MiCollab AM.
 - d** Click **OK**.
- 9** Continue through and complete the configuration. At the end of the configuration, a confirmation dialog box displays. Click **OK**.
- 10** If **MiCollab AM Configuration** does not open automatically after the configuration completes, open **MiCollab AM Configuration**, and select the **Lines** tab.
- 11** In the table from the **Lines** tab, configure callouts for the application. For information on configuring callout settings, see the topic *Configuring Callout Settings*, in the online help system.
- 12** Click **OK** to save all changes.

Completing the Integration

Now you are ready to finish installing MiCollab AM. See *System Installation Guide*, *System Administration Guide* or refer to the *MiCollab AM online help system*, for instructions. For general information on integrations, you may also wish to consult Integrating MiCollab AM with the Telephone System, in the System Installation Guide, and the topic Integrate the System Server with the telephone system, in the online help system.

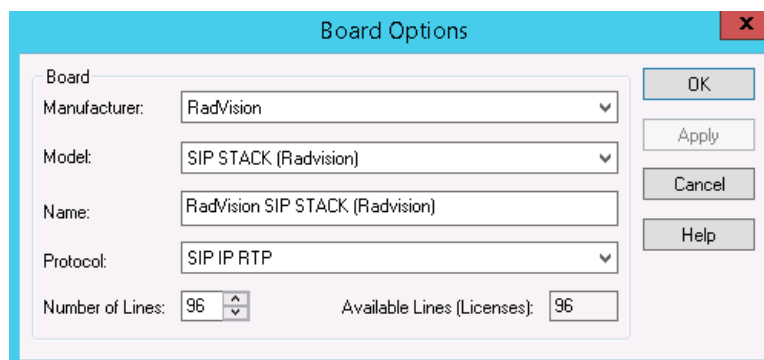
Configuring Existing MiCollab AM for the Integration

To configure exiting MiCollab AM for the telephone integration:

- 1 Open **MiCollab AM Configuration**, and go to the **Main** tab.
- 2 In the **Main** tab, click **Shutdown** to stop the system. Wait until the **Current Status** shows **Stopped**.

NOTE If you have not configured the virtual board with your MiCollab AM system yet, complete **Step 3**. If your MiCollab AM already has the virtual board configured, skip to **Step 4**.

- 3 **[Optional]** Select the **Board** tab, and then click the **Add** button. The **Board** dialog box displays.



- a From the **Manufacturer** dropdown list, select **RadVision**.
 - b From the **Model** dropdown list, select **SIP STACK**.
 - c In the **Name** field, the name for this board is automatically generated. Enter a new name if necessary.
 - d From the **Protocol** dropdown list, select **SIP IP RTP**.
 - e In the **Number of Lines** field, enter the number of lines this board uses. The total number of lines is limited by the capacity of the board and the number of **Available Line Licenses**.
 - f Click **OK**.
- 4 Select the **Switch** tab and click the **Add** button. The **Switch Integration Data Setup** dialog box displays.
 - a From the **Manufacturer** dropdown list, select **Verizon**.
 - b From the **Model** dropdown list, select **HIPC**.
 - c From the **Integration Type** dropdown list, select **SIP Trunk**.

- 5 Click **OK**. The **Switch Options** dialog box displays.

NOTE The settings related to the telephone system in the **Switch Options** dialog box are filled in automatically when you select the correct telephone system during setup.

- 6 If necessary, make any changes to the default settings your site requires in the **Switch Options** dialog box.

NOTE If you need to customize settings on the **Switch Options** dialog box to meet requirements specific to your site, refer to the documentation accompanying the telephone system, the online help, and the guide, *System Installation Guide*.

- 7 Click **OK**. The **Integration Options** dialog box displays.

- 8 In the **Integration Options** dialog box, configure the following options:

- a In the **Local Integration Settings** section, select the **Required Parameter** view and configure the following options:

Table 5. Required Parameters Settings – Integration Options

Field	Value
SIP Server Address	If...
	Then...
	The SIP Routing Manager is not installed
SIP Server Port	Enter the Verizon HIPC IP address.
	The SIP Routing Manager is installed
	Enter the SIP Routing Manager IP address.
SIP Server Port	If...
	Then...
	The SIP Routing Manager is not installed
Transport for outgoing SIP messages	Enter the listen port configured on the Verizon HIPC.
	The SIP Routing Manager is installed
	Enter the listen port configured on the SIP Routing Manager.
Use DNS Discovery Process	Enter UDP, the default.
Local IP Address to bind on	Select this box to use DNS discovery.
SIP Local Connection Port	Select the IP address of the NIC on the Call Server platform that connects to the Verizon HIPC.
SIP Parser Qualifier String	Enter the TCP port MiCollab AM listens for incoming SIP messages. The default port is 5060.
SIP Parser Qualifier String	In cases of a single SIP integration on the call server, enter the local IP address to which the integration is bound. This field is

used by MiCollab AM to match SIP packets to the appropriate SIP integration.

In cases where there are multiple SIP integrations on the call server, use a string that is unique to each SIP integration.

For example:

The extension that will be used as the hunt number on the PBX followed by the "@" symbol and the IP of the call server, such as 5000@172.16.4.202.

The hunt number must be unique across all IP integrations.

The Fully Qualified Domain Name (FQDN) of the switch, such as pbx1.sipdomain.com.

NOTE This setting must match a string in the SIP header that is unique to this particular integration.

b Click **OK**. The **Switch Section Options** dialog box displays.

9 In the **Switch Section Options** dialog box, configure the following options:

a In the **Local Switch Section Settings** section, select the **Required Parameters** view.

b In the **Incoming Hunt Mode** field, select the mode for this integration.

NOTE This integration supports **terminal**, **circular**, **reverse terminal**, and **reverse circular** hunt modes only.

c In the **Hunt Group Access Code** field, enter the Trunk Group Access Code. This number must match with the Directory Number that Verizon has specified as the trunk access number for MiCollab AM.

d Click **OK**.

10 In **MiCollab AM Configuration**, verify that that the telephone system is properly added and configured in the **Switches**, **Switch Sections**, and **Integrations** tabs.

11 Select the **Lines** tab.

12 In the table from the **Lines** tab, configure callouts for the application. For information on configuring callout settings, see the topic *Configuring Callout Settings*, in the online help system.

13 Click **OK** to save all changes.

Configuring MiCollab AM for SIP Failover

MiCollab AM can be configured for automatic failover to the secondary SIP server in the event of the primary/host SIP server failure. Use the instructions provided in this section to add or remove secondary SIP server(s) for failover.

To add a SIP failover server:

- 1 From **MiCollab AM Configuration**, click the **Integrations** tab.
- 2 From the **Integrations** list, select your integration, and then click **Edit**.
- 3 In the **Integration Options** dialog box, go to the **Local Integration Settings** section.
- 4 From the **View** dropdown list, select **Failover Server Settings**.
- 5 Click the **Add Failover Server** button. Two new rows are added to configure the secondary SIP server.
- 6 In the **Secondary SIP Server Address** and **Secondary SIP Server Port** rows, enter the appropriate value as follows:

Table 6. Secondary SIP Server Address and the Secondary SIP Server Port example

Field	Value
Secondary SIP Server Address	<div>Enter the TCP/IP address or an FQDN of the secondary node.</div> <div>For example: The IP address 123.45.6.789 as displayed on the Review/Modify SIP Gateway screen.</div> <div>NOTE This integration requires the machine name to be a fully qualified domain name. Therefore, use the Machine Name field as displayed on the Review/Modify SIP Gateway screen during the integration process.</div> <div>IMPORTANT This value must match the configuration on the Gateway of the secondary node.</div>
Secondary SIP Server Port	Enter the port number of the secondary node. The default value is 5060 .

- 7 From the **View** dropdown list, select **Integration Specific Parameters**. The **Integration Specific Parameters** view displays.
- 8 In the **Integration Specific Parameters** list, enter the information as shown in the following table:

NOTE The parameters in the following table is listed in alphabetical order. The actual Integration Specific Parameters on your system may not be listed in the same order presented in the table below.

Table 7. Integration Specific Parameters

Field	Value
Enable SIP server failover	Select this check box to allow for failover and to enable the failover server setting changes.
Delay (in ms) between Failover attempts	The delay in milliseconds before MiCollab AM attempts to register its port with the SIP server. The default is 1000 ms.
Incoming off hook delay	800
Outgoing off hook delay	0
On hook delay	300
Type of Call Progress to use for External Calls	<p>How this should be set depends on the gateway used for the integration.</p> <ul style="list-style-type: none"> • If the gateway supports call progress through to the endpoint, set to Digital. • If the gateway reports early that the call is connected, such as before the phone rings or while the phone is ringing, set to Media.

- 9 Click **Apply** to save the changes.
- 10 To add another failover server repeat **Steps 4-9**.
- 11 Click **OK** to close the **Integration Options** dialog box.

To remove a SIP Failover Server:

- 1 From **MiCollab AM Configuration**, click the **Integrations** tab.
- 2 From the **Integrations** list, select your integration, and then click **Edit**.
- 3 In the **Integration Options** dialog box, go to the **Local Integration Settings** section.
- 4 From the **View** dropdown list, select **Failover Server Settings**.
- 5 In the **Failover Server Settings** view, click the **Remove Failover Server** button.
- 6 At the confirmation prompt, click **Yes** to confirm the deletion.

NOTE If multiple servers are listed, the last server address and port pair on the list is deleted first.

- 7 Click **Apply** to save the changes, and then click **OK** to close the **Integration Options** dialog box.

Changing the Network Binding Order on the MiCollab AM Platform

If your MiCollab AM server platform is a component of two or more local or wide area networks (LANs or WANs), you must make sure that this integration does not interfere with the normal network operation of the server. By default, MiCollab AM uses the primary (public) network interface card (NIC) in the platform, the first NIC in the network binding order. If you want MiCollab AM to use a NIC other than the first one, you must make several required configuration changes. It is much easier to configure the Integration to use another NIC by simply setting the integration parameter "Local IP Address to bind on" to the address of the NIC card connected to the PBX.

NOTE The operating system gives precedence to the first network connection in the list followed by the remaining connections based on their position in the list.

The instructions in this section ensure that the binding order is correct when you set up the integration. However, if you replace a NIC on the MiCollab AM server platform later, the platform's operating system registers the new adapter at the bottom of its binding order. Restoring the original binding order should correct any problems caused by the change.

IMPORTANT The following procedure shifts the binding order of the network interface cards. To determine which NIC is associated with a specific network connection, right-click the connection in the Network Connections window, and then select **Properties**.

Windows Server 2008 R2 with Service Pack 1

To change the binding order of multiple NICs:

- 8** From the taskbar, click **Start > Control Panel**.
- 9** In the **Control Panel**, click **Network and Sharing Center**.
- 10** On the left pane, select **Change Adapter Settings**.
- 11** Press **Alt** to display the menu bar.
- 12** On the menu bar, select **Advanced**, and then click **Advanced Settings**.
- 13** On the **Adapters and Bindings** tab of **Advanced Settings**, click the network connection that serves MiCollab AM.
- 14** Click the up arrow button to the right of the **Connections** list as many times as needed to move the connection to the top of the list.
- 15** Click **OK**, and then close the **Network Connections** window and the **Control Panel**.

Windows Server 2012 R2

To change the binding order of multiple NICs:

- 1 From the taskbar, click **Start > Control Panel**.
- 2 In the **Control Panel**, click **Network and Sharing Center**.
- 3 On the left pane, select **Change Adapter Settings**.
- 4 Press **Alt** to display the menu bar.
- 5 On the menu bar, select **Advanced**, and then click **Advanced Settings**.
- 6 On the **Adapters and Bindings** tab of **Advanced Settings**, click the network connection that serves MiCollab AM.
- 7 Click the up arrow button to the right of the **Connections** list as many times as needed to move the connection to the top of the list.
- 8 Click **OK**, and then close the **Network Connections** window and the **Control Panel**.

Configuring Quality of Service (QoS)

As of version 6.0, MiCollab AM has no internal support for QoS. QoS must now be implemented externally via group policies as Policy-Based QoS. Refer to your operating system's documentation for details.

Table 8. QoS Configuration

Field	Setting
Application Name	At_TelephonyServer.exe
Protocol	Match the setting used for the integration UDP or TCP
Source Port	<p>MiCollab AM requires a range of ports for audio support. The MiCollab AM audio ports start at the Local Media Base UDP Port configured in the Server tab. Each MiCollab AM line reserves 10 ports. Hence, the port range starts from the number configured there, and goes to the last port of the last line. The formula for calculating the highest port number in the range is as follows:</p> $\text{BasePortNumber} + (\text{NumberOfCXPorts} * 10) - 1.$ <p>Hence, if the base port is 10000, and MiCollab AM has 8 lines, then the port range to use would be:</p> <p>10000:10079</p>
DSCP Value	46