

Application Note: AudioCodes MP-202 Adapter Configuration Guidelines For SIP Trunk Inc MaxCS Release 7.0 Update 1

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Intended audience: AltiGen Authorized Partners



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These instructions apply only to systems using AltiGen-provided SIP Trunks via SIP Trunk Inc.

IMPORTANT: AltiGen does not provide general configuration support for AudioCodes products. AltiGen only supports the MP-202 device as a fax adapter. A support agreement with AudioCodes is required.

The instructions in this guide are based upon configuration for an AudioCodes MP-202 device with firmware version 2.6.4_build_9_1.

Prerequisites

ALTIGEN

In order to implement fax support, your environment must meet the following requirements:

- You must have an AudioCodes MP-202 Telephone adapter
- You will need to provision an additional SIP Trunk group for the adapter
- You must have a dedicated public IP address assigned directly to the MP-202 adapter
- You must have a SIP trunk with T.38 fax support

Audio Codes MP-202 Telephone Adapter Configuration

Complete these steps to configure the MP-202 device. These instructions assume that you are configuring the device for the first time.

1. Open the AudioCodes MP-202 configuration tool. The first page will prompt you to select a language. Choose a language and click **OK**.

🗋 MP20X Management Con: 🗙					
	des			MP-	202 B 2FXS
	Relcome to Your MP20X				
	To conveniently access MP20	19X Management Console. der to access MP20X Management Co IX Management Console you can press ookies are enabled in the browser. To	CTRL+D to add it to your 'Favorite	:s'. frences->Privacy->Cookie	5.
		Please select lar	age for MP20X Management Cor	nsole:	
		Language:	EN English	*	
			√ OK		
tarata barra barra tarata ang Dabata debarra tarata ang					

Figure 1: The AudioCodes Configuration Page

Enter the user name admin and enter a password for this configuration tool. Click OK.



2. On the left, select Network Connections.



Figure 2: Choose Network Connections on the menu

- 3. In the *Action* column, click the **Edit** icon in the row for the WAN interface. You will always use the WAN interface for the SIP trunk network.
- 4. Click Settings.
- 5. Configure the WAN Ethernet properties. You will set the WAN link to static by selecting settings and changing from the default value of DHCP to static address.
 - Enter the public IP address, Default gateway address, and the subnet mask.
 - Specify which DNS server should be used. (The figure shows a cloud DNS IP address of 4.2.2.1).



	WAN	Etherr	net Pr	operties

	-	
		Disable
Name:		WAN Ethernet
Device Name:		eth0
Status:		Connected
Network:		WAN
Connection Type:		Ethernet
MAC Address:		00:90:8f:2e:70:7c
IP Address:		23.24.131.85
Subnet Mask:		255.255.255.248
Default Gateway:		23.24.131.86
DNS Server:		4.2.2.1 208.89.113.142
IP Address Distribution:		Disabled
Received Packets:		1638517
Sent Packets:		16872
Time Span:		266:56:43
	✓ ОК ТАрріу	X Cancel Settings

Figure 3: Configure the WAN properties

- 6. Click Apply. Click OK.
- 7. In the menu on the left, select **Voice Over IP**.
- 8. Click Advanced.
- 9. On the *Signaling Protocol* tab, configure the following settings:
 - Set SIP Transport Protocol to UDP.
 - Check the Use SIP Proxy checkbox.
 - For the *Proxy IP Address or Host Name*, enter the SIP Server IP Address that is provided for you on the *General* tab of the system in the Cloud Order portal.
 - Clear the Use SIP Proxy IP and Port for Registration checkbox.
 - Set SIP Security to Allow All SIP traffic.



	Signaling Protocol	
	Signaling Protocol:	SIP
	SIP Transport Protocol:	UDP -
	Local SIP Port:	5060
	Gateway Name - User Domain:	gw1.sip.us
	Enable PRACK	
	☑ Include ptime in SDP	
	Enable rport	
	Connect media on 180	
	Enable Keep Alive using OPTIONS	
	SIP Proxy and Registrar	
P	Use SIP Proxy	
	Proxy IP Address or Host Name:	
	Proxy Port:	5060
	Maximum Number of Authentication Retries:	10
P	Use SIP Proxy IP and Port for Registration	
	Sip Security:	Allow All SIP traffic
	Use SIP Registrar	
	Use SIP Outbound Proxy	

Figure 4: The Signaling Protocol tab

- 10. Click Apply. Click OK.
- 11. Along the top, select the **Media Streaming** tab.



12. In the Codecs section, set codec 1 to G.711 kbps u-Law. Set Packetization Time to 20 for all codecs.

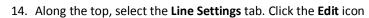
Codecs			
Codecs Priority	Supported Codecs	Packetization Time	
1st Codec	G.711, 64kbps, u-Law 🔻	20 🔻	
2nd Codec	G.711, 64kbps, u-Law 🔻	20 🔻	
3rd Codec	G.711, 64kbps, u-Law 🔻	20 🔻	
4th Codec	G.711, 64kbps, u-Law 💌	20 🔻	

Figure 5: Set Codec parameters

13. On the top, select the **Voice and Fax** tab. Set *Image Data Redundancy Level* to 3. Set *T30 Control Data Redundancy Level* to 3. Make sure that your configuration matches all of the settings in the next figure.

Gain Control	
Enable Automatic Gain Control	
Jitter Buffer	
Minimum Delay (10 to 150 milliseconds):	35 milliseconds
Optimization Factor (1 to 13):	7
Silence Compression	
Enable Silence Compression	
Echo Cancellation	
✓ Enable Echo Cancellation	
Fax and Modem Settings	
Fax Transport Mode:	T.38 Relay 🔻
Max Rate:	14.4 Kbps 🔻
Max Buffer:	1024
Max Datagram:	320
Image Data Redundancy Level:	3
T30 Control Data Redundancy Level:	3
Fax Relay Jitter Buffer Delay:	0
Error Correction Mode	
Modem Transport Mode:	Bypass 🔻
Modem Bypass Payload Type:	103
Fax/Modem Bypass Codec:	G.711, 64kbps, u-Law 💌
Enable CNG Detection	

Figure 6: The Fax and Modem Settings



on the row that you want to change.

15. Log into the MAXCS Cloud order portal and retrieve the Fax details that are provided on the *General* tab of your system.

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16. Add the telephone number:

📥 A L T I G E N

- User ID is the telephone number
- Display Name is the same telephone number that you configured in the user ID
- Authentication User Name is the SIPTrunk username from the General tab in the order portal.
- Authentication Password is the SIPTrunk password from the General tab in the order portal.

Dine Settings			
Line Number:	1		
User ID:	14087140267		
Block Caller ID			
Display Name:	14087140267		
SIP Proxy			
Authentication User Name:	5232798393		
Authentication Password:	•••••		
Advanced Line Parameters			
Line Voice Volume (-31 to +31 db):	0		
Enable Supplementary Services			

Figure 7: The Line Settings tab

17. On the left menu, select **Advanced**. Click the **Remote Administration** icon to configure remote login access.



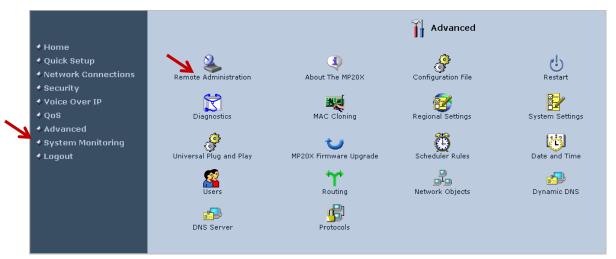


Figure 8: Configure remote login access

18. Check the boxes as shown in the next figure and click **Apply**.

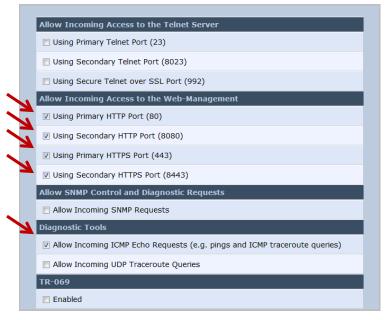


Figure 9: Access parameters



- 19. The last step is to change the authentication method to IP Authentication in SIP Trunk Inc. Log onto https://partner.altigen.com and open a support case to request this change. In the case, include the following information:
 - The public IP address assigned to the MP-202 device
 - The order ID number from the associated account in the Cloud portal, along with the system name

AltiGen Technical Support

AltiGen does not provide general configuration support for AudioCodes products. For general configuration information, refer to your AudioCodes documentation.

AltiGen provides technical support to Authorized AltiGen Partners and distributors only. End user customers, please contact your Authorized AltiGen Partner for technical support.

Authorized AltiGen Partners and distributors may contact AltiGen technical support by the following methods:

- You may request technical support on AltiGen's Partner web site, at https://partner.altigen.com. Open a case on this site; a Technical Support representative will respond within one business day.
- Call 888-ALTIGEN, option 5, or 408-597-9000, option 5, and follow the prompts. Your call will be answered by
 one of AltiGen's Technical Support Representatives or routed to the Technical Support Message Center if no
 one is available to answer your call.

Technical support hours are 5:00 a.m. to 5:00 p.m., PT, Monday through Friday, except holidays.

If all representatives are busy, your call will be returned in the order it was received, within four hours under normal circumstances. Outside AltiGen business hours, only urgent calls will be returned on the same day (within one hour). Non-urgent calls will be returned on the next business day.

Please be ready to supply the following information:

- Partner ID
- AltiGen Certified Engineer ID
- Product serial number
- AltiWare or MAXCS version number
- Number and types of boards in the system
- Server model
- The telephone number where you can be reached

AltiGen Communications, Inc. 659 River Oaks Parkway San Jose, CA 95134 Telephone: 888-AltiGen (258-4436) Fax: 408-597-9020 E-mail: info@altigen.com Web site: www.altigen.com

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