

Avaya Solution & Interoperability Test Lab

# **Application Notes for Configuring XMedius XM Fax Software with Avaya IP Office 11 via a SIP Trunk - Issue 1.0**

#### Abstract

These Application Notes describe the procedures for configuring the XMedius XM Fax Software with Avaya IP Office using a SIP trunk interface.

XMedius XM Fax is fax software that sends and receives fax calls over an IP network. In the tested configuration, XMedius XM Fax interoperated with Avaya IP Office to send/receive faxes using SIP trunk facilities.

Readers should pay attention to **Section 2**, in particular the scope of testing as outlined in **Section 2.1** as well as the observations noted in **Section 2.2**, to ensure that their own use cases are adequately covered by this scope and results.

Information in these Application Notes has been obtained through DevConnect compliance testing and additional technical discussions. Testing was conducted via the DevConnect Program at the Avaya Solution and Interoperability Test Lab.

# 1. Introduction

These Application Notes describe the procedures for configuring XMedius XM Fax with Avaya IP Office (IPO) using SIP trunks.

XMedius XM Fax is host-based Fax over IP that uses XMedius Fax driver to send and receive fax calls over an IP network. In the tested configuration, XM Fax establishes a SIP trunk interface with T.38 fax from Avaya IP Office to send and receive fax.

In the compliance testing, Avaya IP Office Server Edition system consists of Avaya IP Office Primary Linux running on Virtualized Environment and a 500V2 Expansion.

## 2. General Test Approach and Test Results

The feature test cases were performed manually. Internal and external fax calls to and from XM Fax server were made. The fax calls were sent and received using the web interface of XM Fax and the analog fax destination at the local and remote sites.

The serviceability test cases were performed manually by disconnecting/reconnecting the Ethernet cable to the XM Fax server, busyout and release SIP the trunk and by rebooting the XM Fax server.

DevConnect Compliance Testing is conducted jointly by Avaya and DevConnect members. The jointly-defined test plan focuses on exercising APIs and/or standards-based interfaces pertinent to the interoperability of the tested products and their functionalities. DevConnect Compliance Testing is not intended to substitute full product performance or feature testing performed by DevConnect members, nor is it to be construed as an endorsement by Avaya of the suitability or completeness of a DevConnect member's solution.

Avaya recommends our customers implement Avaya solutions using appropriate security and encryption capabilities enabled by our products. The testing referenced in this DevConnect Application Note included the enablement of supported encryption capabilities in the Avaya products. Readers should consult the appropriate Avaya product documentation for further information regarding security and encryption capabilities supported by those Avaya products.

Support for these security and encryption capabilities in any non-Avaya solution component is the responsibility of each individual vendor. Readers should consult the appropriate vendor-supplied product documentation for more information regarding those products.

For the testing associated with these Application Notes, the interface between Avaya systems and the XMedius XM Fax do not utilize TLS and secure media SRTP encryption features as requested by XMedius.

#### 2.1. Interoperability Compliance Testing

The interoperability compliance test included feature and serviceability testing. The feature testing focused on verifying the following with the XM Fax:

- Proper handling of faxes via the SIP trunk including send/receive, internal fax, external fax over SIP trunk simultaneous bi-directional faxes, and miscellaneous failure scenarios.
- Proper handling of faxes with different pages, complexity, format and data rates.
- No adverse impact on any internal or external calls during faxes.

The serviceability testing focused on verifying the ability of XM Fax to recover from adverse conditions, such as disconnecting/reconnecting the Ethernet cable to the XM Fax server.

#### 2.2. Test Results

XMedius XM Fax successfully passed all compliance testing.

#### 2.3. Support

North American Technical support for XM Fax Software can be obtained by contacting XMedius at.

- North America: +1-866-615-3066
- +1 514-787-2122
- EMEA: +33 (0) 1 70 92 13 12
- Email: support.software@xmedius.com
- Website: <u>https://support.xmediusfax.com</u>

# 3. Reference Configuration

The test configuration was designed to emulate a local site and a remote site. **Figure 1** illustrates the configuration used in these Application Notes.

In the compliance testing, the Avaya IP Office Server Edition system consists of Avaya IP Office Primary Linux running on Virtualized Environment and a 500V2 Expansion. The IPO Primary was configured to connect to PSTN via a SIP trunk while the 500V2 Expansion connected to PSTN via a PRI trunk, IPO Primary and 500V2 Expansion is communicated by Small Community Network (SCN) IP Office Line. The following are typical scenarios verified during the compliance test:

- Bi-directional faxed between XM Fax server and the local fax endpoint 1 that connected to an analog port in the IPO 500V2 Expansion.
- Bi-directional faxes between XM Fax server and PSTN fax endpoint 2 via SIP Service Provider.



# Figure 1: XMedius XM Fax interoperating with Avaya IP Office via SIP Trunk

# 4. Equipment and Software Validated

The following equipment and software/firmware were used for the sample configuration provided:

Equipment	Release/Version
Avaya IP Office Primary Linux running on	11.0.4.1 Build 11
Virtualized Environment	
Avaya IP Office 500V2 Expansion	11.0.4.1 Build 11
Avaya IP Office Manager	11.0.4.1 Build 11
Avaya 1140E SIP Deskphones	4.04.23
Avaya 96x1 IP Deskphones	6.8
XMedius XM Fax Software running on	9.0.0.510 with XMFaxDriver 9.0.0.526
Microsoft Windows 2016 Server	

**Note:** Testing was performed with Avaya IP Office Server Edition Solution that requires an Expansion IP Office 500 V2 to support analog used by fax endpoint. Compliance Testing is applicable when the tested solution is deployed with a standalone IP Office 500.

# 5. Configure Avaya IP Office

This section provides the procedures for configuring Avaya IP Office. The procedures include the following areas:

- Verify Avaya IP Office License
- Obtain LAN IP address
- Enable SIP trunks
- Administer SIP line
- Administer Incoming Call Route
- Administer Short Code
- Administer IP Office Line

#### 5.1. Verify Avaya IP Office License

From a PC running the Avaya IP Office Manager application, select **Start**  $\rightarrow$  **Programs**  $\rightarrow$  **IP Office**  $\rightarrow$  **Manager** to launch the Manager application. Select the correct IP Office system and log in with the appropriate credentials.

The **Avaya IP Office R11 Manager** screen is displayed. From the configuration tree in the left pane, select **License**. Verify that the **SIP Trunk Channels** license is "Valid", and that the **Instances** value is sufficient for the desired maximum number of simultaneous faxes.

Configuration					$ X  \vee   <$
BOOTP (7)	Licence Remote Server				
🥙 Operator (3) Solution					
	Feature	Instances	Status	Expiry Date	Source
🖬 📲 Group(2)	Mobile Worker	384	Obsolete	Never	PLDST
Short Code(14)	Office Worker	384	Valid	Never	PLDST
Time Profile(0)	Avaya Softphone Licence	100	Valid	Never	PLDST
	VMPro TTS (Scansoft)	40	Obsolete	Never	PLDST
📲 User Rights(4)	VMPro TTS Professional	40	Valid	Never	PLDSI
	IPSec Tunnelling	10	Obsolete	Never	PLDSI
POSE110	Power User	384	Valid	Never	PLDSI
≣™™ System (1) ≣™filline (6)	Customer Service Agent	100	Valid	Never	PLDSI
	Customer Service Supervisor	100	Valid	Never	PLDSI
🖶 🛷 Extension (18)	Avaya IP endpoints	384	Valid	Never	PLDSI
🖶 📲 User (23)	IP500 Voice Networking Channels	32	Obsolete	Never	PLDST
Group (2)     Short Code (72)	SIP Trunk Channels	1024	Valid	Never	PLDSI
Service (0)	IP500 Universal PRI (Additional cha	100	Obsolete	Never	PLDST
Incoming Call Route (8)	CTI Link Pro	5	Valid	Never	PLDST
- A Directory (0)	Wave User	16	Obsolete	Never	PLDST
	3rd Party IP Endpoints	384	Valid	Never	PLDST
IP Route (1)	Centralized Endpoints	100	Obsolete	Never	PLDST
Account Code (U)	Essential Edition	5	Obsolete	Never	PLDST
Handle User Rights (13)	RR+ Dreferred Edition (VM Dro)	5	Obsolete	Novar	DINGI
⊕ <b>*</b> ARS (1)	<				
- 🯧 Location (0)	<				
🔤 🥁 Authorization Code (0) Mag QTIPO				<u>O</u> K <u>C</u> an	cel <u>H</u> el

#### 5.2. Obtain LAN IP Address

From the configuration tree in the left pane, select **System** to display the **System** screen for the **IPOSE110** in the right pane. Select the **LAN1** tab, followed by the **LAN Settings** sub-tab in the right pane. Make a note of the **IP Address**, which will be used later to configure the Dialogic FDTool. Note that IP Office can support SIP trunks on the LAN1 and/or LAN2 interfaces. The compliance testing used the LAN1 interface.

Configuration	12	IPOSE110*	📸 - 🔤   🗙   🗸   <   >
BOOTP (7) ⊕ - Ø Operator (3) ⊕ - Ø Operator (3)	System LAN1 LAN2 DNS LAN Settings VolP Networ	Voicemail Telephony Directory Services	System Events SMTP S • •
	IP Address IP Mask	10     .     10     .     97     .     110       255     .     255     .     255     .     192	
User Rights(4) Gradient Costion(0) Gradient Costion(1) Gradient Costing Cos	Number Of DHCP IP Addresses DHCP Mode O Server O Client O Dis	16 🔹 abled Advanc	ed

#### 5.3. Enable SIP Trunks

Select the VoIP sub-tab. Ensure that SIP Trunks Enable is checked as shown below.

<b>1</b>		IPOS	E110*					<b>e</b> ř	- 🗐   🗙   -	✓   <
System LAN1 LAN2 DNS	Voicemail	Telephony	Directory Se	rvices S	ystem Events	SMTP	SMDR	VoIP	VoIP Security	Con •
LAN Settings VolP Network	Topology									
🖂 H323 Gatekeeper Enable —										^
🗌 Auto-create Extn	_ At	uto-create U	ser		🗌 H323 Rem	ote Extn	Enable			
H.323 Signalling over TLS	Disabled	~	,		Remote Call	Signalling	Port 17	20	A V	
SIP Trunks Enable										
🗹 SIP Registrar Enable										
Auto-create Extn/User						C	SIP Ren	note Ext	tn Enable	
SIP Domain Name	ipocc.c	om								
SIP Registrar FQDN										
	🗹 UDF	)	UDP Port	5060	• •	Ren	note UDP	Port 5	060	*
Layer 4 Protocol	🗹 ТСР	I	TCP Port	5060	▲ ▼	Ren	note TCP F	ort 5	060	*
	🗹 TLS		TLS Port	5061	▲ ▼	Ren	note TLS P	ort 5	061	*
Challenge Expiry Time (secs)	10	•								
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							<u>0</u> K		<u>C</u> ancel	<u>H</u> elp
										F

#### 5.4. Administer SIP Line

From the configuration tree in the left pane, right-click on Line and select New  $\rightarrow$  SIP Line from the pop-up list to add a new SIP line. Select the SIP Line displays in the right pane. Select SIP Line tab and provide the following values for the fields below.

- Line Number: enter an unassigned SIP Line number in the IPO system
- ITSP Domain Name: enter the IP address of XM Fax server
- Local Domain Name: enter the LAN1 IP address of IPO
- In Service: check the checkbox to place the SIP Line in service
- **Check OOS**: tick the checkbox to enable IPO sending out OPTIONS to the XM Fax server to check the status of SIP Line

Leave other fields at default values.

Configuration	E	SIP Line - Line 8		📸 - 🔛   🗙   🗸   <   >
⊞ 👫 BOOTP (2)	SIP Line Transport Call Details VolP	SIP Credentials SIP Advanced Engineering		
Operator (3)     Solution				A
internation inter	Line Number	8	In Service	
🕀 🖓 Group(1)	ITSP Domain Name	10.33.1.60	Check OOS	
Short Code(46)		10.22.1.110		
Directory(0)     Time Desclar(0)	Local Domain Name	10.33.1.110		
Account Code(0)	URI Type	SIP URI	Session Timers	
I User Rights(11)	La antina	Cloud	Refresh Method	Auto
Location(2)	Location	Cioud	nen est men ed	
Prosentaria (1)			Timer (sec)	On Demand
● System (1)				
1	Prefix			
	National Prefix	0		
	National Frenk	č		
9	International Prefix	00		
10	Country Code		Redirect and Transfer	
E Control Unit (9)			Incoming Supervised REFER	Auto
Extension (16)	Name Priority	System Default V		
Group (0)	Description		Outgoing Supervised REFER	Auto
H Short Code (8)			Send 302 Moved Temporarily	
Service (0)			Outgoing Blind REFER	
Incoming Call Route (5)				
License (33)	4		- 1	× ×
🗄 🕂 🖌 ARS (1)				
🗄 🔤 Location (2) 🗸 🗸			<u>O</u> K	Cancel <u>H</u> elp
< >				

Select the **Transport** tab and provide the following configuration for the fields:

- **ITSP Proxy Address**: enter the IP address of XM Fax server
- Network Configuration Layer 4 Protocol: select *UDP* from this list
- Network Configuration Send Port: enter the port 5060
- Network Configuration Listen Port: enter the port 5060
- Network Configuration Use Network Topology Info: select LAN1 from the list

Retain the default values for the remaining fields.

Configuration	SIP Line - Line 8	📸 • 🔛   🗙   🗸   <   >
	SIP Line       Transport       Call Details       VoIP       SIP Credentials       SIP Advanced       Engineering         ITSP Proxy       Address       10.33.1.60       Intervention       Send Port       5060         Network       Configuration       VDP       Send Port       5060         Use       Network       Topology       Info       LAN 1       Listen Port       5060         Explicit       DNS       Server(s)       0       0       0       0       0       C         Calls       Route via       Registrar       Separate       Separate <t< td=""><td></td></t<>	
🗉 🚋 Location (2)	<	>
Authorization Code (0)	ŌK	<u>C</u> ancel <u>H</u> elp
Ready		III .::

Select the **Call Details** tab, and click **Add** button (not shown) to display the **SIP URI** pop-up window. Enter the SIP line number that is created above for **Incoming Group** and **Outgoing Group**. Set **Max Calls per Channel** to the desired maximum number of simultaneous faxes allowed, in this case "10". Retain the default values in the remaining fields.

📶 SIP Line - 8   Call D	Details   SIP URI				×			
New URI								
Incoming Group 🛛 🗸 Max Sessions 10								
Outgoing Group 8 V								
Credentials 0: <1	None> ~							
	Display	Content	Field meaning					
			Outgoing Calls	Forwarding/Twinning	Incoming Calls			
Local URI	Auto ~	Auto ~	Caller ~	Original Caller 🗸 🗸	Called $\lor$			
Contact	Auto 🗸	Auto ~	Caller ~	Original Caller $\checkmark$	Called $\checkmark$			
P Asserted ID	Auto 🗸	Auto 🗸	Caller ~	Original Caller $\checkmark$	Called $\checkmark$			
P Preferred ID	None 🗸	None 🗸	None 🗸	None 🗸	None $\vee$			
Diversion Header	None 🗸	None 🗸	None $\lor$	None $\vee$	None $\vee$			
Remote Party ID	None 🗸	None 🗸	None $\vee$	None	None $\vee$			
				ОК	Cancel Help			

The screen below shows the SIP URI is created in the Call Details of SIP Line – Line 8.

Configuration	SIP Line - Line 8	📑 - 🔄   🗙   🗸   <   >
Short Code(46)	SIP Line Transport Call Details VoIP SIP Credentials SIP Advanced Engineering	
Directory(0)     Time Profile(0)	SIP URIs	
Account Code(0)	URI Groups Credential Local URI Contact P Asserted ID P Preferred ID	Diversion Heade Add
Eccation(2)	1 8 8 0: <none> Auto Auto Auto</none>	
in the sector (1)		Remove
IPOSE110		Edit
□(		
	<	>
	SIP Line Appearances	
> 10	Line ID Incoming ID Outgoing ID Groups Credential Local URL Contact	P Asserted ID
Extension (16)		Addin
User (20)   Group (0)		Remove
Short Code (8)		Edit
Service (0)		
IP Route (4)		
License (33)	<	>
Elocation (2)		
Authorization Code (0)	OK	Cancel Help
< >		

KP; Reviewed: SPOC 4/1/2020

Select the **VoIP** tab. For **Codec Selection**, select *Custom* and select the applicable G.711ULAW and G.711ALAW codecs in the expanded list. Note that XM Fax supports the G.711 variants.

- Fax Transport Support select *T38 Fallback* from the drop-down list.
- Check *Re-invite Supported*
- **DTMF Support** select *RFC2833/RFC4733* from the drop-down list
- Media Security select *Disabled*

Retain the default values in the remaining fields.

Configuration	12	SIP Line - Line 8*	📸 <b>-</b> 🖭   🗙   🗸   >
Group(1)	SIP Line Transport Call	Details VolP SIP Credentials SIP Advanced Engineering	
Directory(0)			Local Hold Music
Time Profile(0) Account Code(0)			Re-invite Supported
🗄 🏰 User Rights(11)	Codec Selection	Custom	Codec Lockdown
		Unused	Allow Direct Media Path
System (1)		G.722 64K G.729(a) 8K CS-ACELP	Force direct media wit
			PRACK/100rel Supported
2			
		£	
Extension (16)		>>>	
and the short Code (8)	Fax Transport Support	T38 Fallback	~
	DTMF Support	RFC2833/RFC4733	~
	Media Security	Disabled $\checkmark$	
	1		
Location (2)     Authorization Code (0)			
EXP110 V			<u>O</u> K <u>C</u> ancel <u>H</u> elp

Retain all values in the SIP Credentials, SIP Advanced and Engineering tabs as default.

#### 5.5. Administer Incoming Call Route

From the left pane, right-click on **Incoming Call Route**, select **New** (not shown) from the pop-up list to add a new route. For **Line Group ID** in the **Standard** tab, select the incoming group number from **Section 5.4** which corresponds to the SIP line, in this case Line Group ID is *8*.

Configuration		X	8	C	🛉 📲   🗙   🗸   <   >
	^	Standard Voice Recording	Destinations		
🕀 🥙 Operator (3)					
Solution				_	
User(37)		Bearer Capability	Any Voice	~	
Bung Group(1)			0		
Directop((0)		Line Group ID	٥ ``	~	
Time Profile(0)		Incoming Number			
Account Code(0)		inconing runner		_	
		Incoming Sub Address			
E				_	
IPOSE110		Incoming CLI			
🗄 🖘 System (1)		Locale		~	
⊞ारि Line (6)		Locale		-	
🗄 🖘 Control Unit (9)		Priority	1 - Low	~	
🗄 🖓 Extension (16)				-	
🖭 📲 User (20)		Tag			
Group (0)		Listd Music Course	Suntain Courses		
		Hold Music Source	System Source	~	
Service (0)		Ring Tone Override	None	~	
incoming Call Route (5)					
7 12172107602					
7 12172107002					
7 12172107075					
9 2063099337					
IP Route (4)					
License (33)		-			
				<u>0</u> K	Cancel <u>H</u> elp
III. in Leastion (7)	Υ.				

Select the **Destinations** tab. For **Destination**, enter "." to match any dialed number from the XM Fax server.

Configuration	XXX		6	📸 🗕 🔛 🛛 🗙 🛛 🖌 🗠 😽
BOOTP (7)	Standa	ard Voice Recording Destinations		
Solution		TimeProfile	Destination	Fallback Extension
🖶 📲 User(29)	•	Default Value	. 🗸	~
🖶 🎇 Group(2)				
🖶 🥬 Short Code(14)				
Directory(0)				
Account Code(0)				
🗄 📲 User Rights(4)				
Location(0)				
POSETIU				
System (1)				
Energy Control Unit (11)				
Extension (18)				
in a User (23)				
🗄 🖓 Group (2)				
🖶 🥬 Short Code (72)				
Service (0)				
🖨 😰 Incoming Call Route (8)				
🔁 2				
- 🕞 б				

#### 5.6. Administer Short Code

From the configuration tree in the left pane, right-click on **Short Code** and select **New** from the popup list to add a new short code. In the compliance testing, users on IP Office are designated with fax numbers 52xx, and the fax calls are routed over the SIP trunk to the XM Fax server.

- Code: enter 52N; as the prefix started from 52
- **Feature:** select **Dial** from the list
- **Telephone Number**: enter 52N
- Line Group ID: enter the outgoing group number *8* from Section 5.4, which corresponds to the SIP line.

Configuration		XXX	52N;: Dial		📥 - 🔤   🗙	✔   <   >	
	^	Short Code					
Short Code(46)		Code	52N;				
Time Profile(0)		Feature	Dial 🗸				
		Telephone Number	52N				
⊞		Line Group ID	8 ~				
		Locale	~				
		Force Account Code					
User (20)     Group (0)		Force Authorization Code					
Short Code (8)							
9× 1N;							
3N; 46N;							
9× 52N; •••9× 53N;							
9× 9N							
Service (0)							
- <b>(b)</b> 8 - <b>(b)</b> 7 12172107602							
7 12172107673	~			<u>O</u> K	<u>C</u> ancel	<u>H</u> elp	
Ready						F¥.	

#### 5.7. Administer IP Office Line

The IP Office Small Community Network (SCN) Lines in the IP Office primary and IP Office expansion were previously created during the setup of IP Office Server Edition system when a 500V2 expansion joined to IP Office Primary.

The picture below shows Line 1 - VoIP Settings in the IP Office Primary for the compliance test. The Fax Transport Support was set to *T.38 Fallback*.

Configuration		📸 🕶   🗙   🖌   <   >	
BOOTP (2)	Line Short Codes VolP S	ettings	
Solution			Out Of Band DTMF ^
⊡ ∰ User(37)			Allow Direct Media I
	Codec Selection	Custom ~	
Directory(0)     Time Profile(0)     Account Code(0)     User Rights(11)     Direction(2)     IPOSE110     I     f Line (6)     f Line (6)     f 1     2     7     8		Unused G.711 ALAW 64K G.722 64K G.722 64K G.722 64K G.722 64K G.722 64K G.722 64K G.722 64K G.722 64K G.722 64K	
10			
	Fax Transport Support	T38 Fallback	~
⊡ 1 User (20) 🙀 Group (0)	Call Initiation Timeout (s)	4	
Bervice (0)	Media Security	Same as System (Preferred) V	

The picture below shows Line 17 - VoIP Settings in the IP Office Expansion for the compliance test with the XM Fax server. The Fax Transport Support was set to *T.38 Fallback*.

Configuration	E	📸 - 🔤   🗙   🗸   <   >	
■ <b>8</b> BOOTP (2)	Line Short Codes VolP S	ettings T38 Fax	
·····································			VolP Silence Suppres
			✓ Out Of Band DTMF
Short Code(46)	Codec Selection	Custom	Allow Direct Media
Directory(0) Time Profile(0)		Unused	
Account Code(0)		G.711 ALAW 64K >>> G.711 ULAW 64K G.722 64K G.729(a) 9K CS-A CELP	
uerraghts(11) uerraghts(11) uerraghts(12)		G.723.1 6K3 MP-MLQ	
IPOSE110			
EXP110		<<<	
□ 17 Line (6)			
2		1	
18			
20	Fax Transport Support	T38 Fallback	~
Entension (27)			
🖳 📲 User (19)	Call Initiation Timeout (s)	* •	
	Media Security	Same as System (Preferred) $\sim$	

# 6. Configure Dialogic XM Fax

This section describes the configuration of XM Fax software.

Launch the **XM Fax – Administration Login** webpage and enter a proper user name and password to login.

Fax	Languag	e: English 🗸
Login OWindows Authentication Fax Server Authentication Administrator name: Password: System Administrator Site Administrator Site Name: Remember me on this comp	Login	
Copyrigh	lutions Inc XM Fax 9.0.0.518	

Select **Driver** icon from the left pane. The **Driver Properties** section displays in the right side of the window. In the **Options** tab, set the **Maximum Number of Channels** and **Preferred Number of Channels** available by XM Fax.

Fax	System Configuration	Sites	System Monitor	Phone Books	
<ul> <li>SERVER3</li> <li>Config Manager</li> <li>Fax Manager</li> <li>Fax Manager</li> <li>Outbound Modification Table</li> <li>Channels</li> <li>Channels</li> <li>Rasterizer</li> <li>Fax Folder Mediation</li> <li>Fax Archive</li> <li>CoConfig</li> <li>Fault Tolerance</li> <li>SMTP Gateway</li> <li>SMTP Gateway</li> <li>Proxy</li> </ul>	Options       FoIP       SIP       SIP Security         Options       Options       24         Log Size (MB):       20       20         Information Logging Level:       Low Le         ✓       Enable Log Archiving         Archive Retention (in days):       11         Debug       Debug	ty H.323	Dial Plan Peer List	t Network Capture	
	FoIP Channel Configuration Maximum Number Of Channels:* Preferred Number Of Channels: *Changes to properties marked with an service is restarted.	24 24 asterisk will ta	ake effect when the	ОК	Cancel

Select the **FoIP** tab, check the **Enable ECM** field and keep other fields at default values.

Fax	System Configuration	Sites	System M	Ionitor	Phone Books	
<ul> <li>✓ SERVER3</li> <li> <sup>(*)</sup> Config Manager         <sup>(*)</sup> Fax Manager         <sup>(*)</sup> Fax Manager         <sup>(*)</sup> Outbound Modification Table         <sup>(*)</sup> Channels         <sup>(*)</sup> Rasterizer         <sup>(*)</sup> Fax Folder Mediation         <sup>(*)</sup> Fax Folder Mediation         <sup>(*)</sup> Fax Archive         <sup>(*)</sup> CoConfig         <sup>(*)</sup> Fault Tolerance         <sup>(*)</sup> SMTP Gateway         <sup>(*)</sup> XML Gateway         <sup>(*)</sup> Proxy</li></ul>	Options       FolP       SIP       SIP Securit         Options       ✓       FolP       SIP       SIP Securit         Options       ✓       Enable ECM*       G         Terminal Resolution Capacity:*       U       U       Binding Interface:*       O.         Call Delay (milliseconds):       0       *       *Changes to properties marked with an service is restarted.	y H.323 roup 3 (1d) tra (400x400) ).0.0 asterisk will ta	Dial Plan	Peer List	Network Capture	
					ок	Cancel

Leave all fields at default values in the **SIP** tab.

Fax	System Configuration Sites Syst	em Monitor Phone Books
<ul> <li>SERVER3</li> <li>Serveration (Section 2)</li> <li>Config Manager</li> <li>Tax Manager</li> <li>Serveration (Section 2)</li> </ul>	Driver Properties	
<ul> <li>Outbound Modification Table</li> <li>Channels</li> <li>Rasterizer</li> <li>Fax Folder Mediation</li> <li>Fax Archive</li> <li>CoConfig</li> <li>CoConfig</li> <li>Fault Tolerance</li> <li>SMTP Gateway</li> <li>XML Gateway</li> <li>Proxy</li> </ul>	General Local SIP UDP Port:* 5060 Local SIP TCP Port:* 5060 Local SIP TLS Port:* 5061 Wait For DTMF Code Input Maximum Delay Between DTMF signals (seconds): Print SIP Messages *Changes to properties marked with an asterisk will take effect service is restarted.	10

Select the **Peer List** tab, click **Add SIP Peer** button to add a SIP Peer for IP Office.

Fax	System Confi	iguration	Sites System Mo	nitor Phone Books
<ul> <li>SERVER3</li> <li>Server3</li> <li>Server3</li> <li>Fax Manager</li> <li>Priver</li> <li>Outbound Modification Table</li> </ul>	Options FolP SIP	SIP Security	H.323 Dial Plan	Peer List Network Capture
<ul> <li>Channels</li> <li>Rasterizer</li> <li>Fax Folder Mediation</li> <li>Fax Archive</li> <li>Fax Archive</li> <li>CoConfig</li> <li>Fault Tolerance</li> <li>SMTP Gateway</li> <li>XML Gateway</li> <li>Proxy</li> </ul>	Peer List           Host Name           server3           10.33.1.12           ✓         10.33.1.110	Protocol SIP SIP SIP	Peer Description Session Manager IP Office	Add SIP Peer Add H.323 Peer Remove Properties
	Default SIP Propertie Default H323 Propert	s ies		

The **Peer Properties** section displays in the right side of the window. In the **General** tab, enter the following parameters:

- Host Name: enter the SIP entity IP address of IP Office 10.33.1.110
- **Peer Description**: enter a description
- **Transport**: select **UDP** port
- **Port**: enter the port *5060*
- Media Type: select T.38 with Fallback to G.711

And keep other fields at default values.

Fax	System Configuration Sites System Monitor Phone Books
<ul> <li>SERVER3</li> <li>SERVER3</li> <li>Config Manager</li> <li>Fax Manager</li> <li>Fax Manager</li> <li>Outbound Modification Table</li> <li>Channels</li> <li>Channels</li> <li>Rasterizer</li> <li>Fax Folder Mediation</li> <li>Fax Folder Mediation</li> <li>Fax Archive</li> <li>CoConfig</li> <li>Fault Tolerance</li> <li>SMTP Gateway</li> <li>SMTP Gateway</li> <li>YML Gateway</li> <li>Proxy</li> </ul>	Peer Properties         General T.38 G.711 Codecs Inbound Modification Table         Options       Options         Host Name:       10.33.1.110         Peer Description:       IP Office         Transport:       UDP V         Port:       5060         Media Type:       T.38 with Fallback to G.711         G.711 fallback delay after fax detection (milliseconds):       3500         Delay Before Call Completion (seconds):       1         Voice Call Timeout (seconds):       40         "user" parameter in SIP URI:       phone V         VIA and CONTACT Headers Host Name Override:       V.34 Enabled         V.29 Enabled       Use Proxy         Host Name:       Disable this peer for incoming calls

Select **T38** tab and leave all fields at default values.

Fax	System Configuration Sites System Monitor Phone Books
<ul> <li>✓ SERVER3</li> <li>○ (◊) Config Manager</li> <li>○ ○ ○ □ □ □ □ □ □ □ □ □ □ □ □ □ □ □ □ □</li></ul>	Peer Properties
<ul> <li>Fax Manager</li> <li>Oriver</li> </ul>	General T.38 G.711 Codecs Inbound Modification Table
<ul> <li>Outbound Modification Table</li> <li>Channels</li> <li>Rasterizer</li> <li>Fax Folder Mediation</li> <li>Fax Archive</li> <li>Fax Archive</li> <li>CoConfig</li> <li>Fault Tolerance</li> <li>SMTP Gateway</li> <li>ML Gateway</li> <li>Proxy</li> </ul>	Options Outbound Initial Media Offer: Audio CNG: Send using RTP Delay Before Re-INVITE (seconds): 2 Leading T.38 "no-signal" Packets: 3 Send T.38 Re-INVITE (Sending Side) Delay Before Re-INVITE (seconds): 2 T38 Redundancy Low Speed Redundancy Depth: 1 High Speed Redundancy Depth: 1 V

Select **G.711** tab and leave all fields at default values.

Fax		System	Config	ration	Sitos	System Monitor	Phone Books	7
✓ SERVER3	Peer Prope	rties	-connyt		51(65	System Promoti	THORE BOOKS	
	General SRTP- Cr	T.38 edia Encryp rypto Suit SRTP Au Use MKI MKI By	G.711 tion 9: AES_ thentication	Codecs CM_128_HN on I: 4 V	Inbound	Modification Table		OK Back

Select **Codecs** tab, in the **Codecs** tab select **Add** button to add a designed codecs to the **Supported Codecs** list. In the compliance test, two codecs *G.711 Mu-law* and *G.711 A-Law* were used.

Fax	System Configuration Sites System Monitor Phone Books	
<ul> <li>SERVER3</li> <li>Config Manager</li> <li>Fax Manager</li> <li>Priver</li> <li>Outbound Modification Table</li> <li>Channels</li> <li>Rasterizer</li> <li>Fax Folder Mediation</li> <li>Fax Archive</li> <li>CoConfig</li> <li>Fault Tolerance</li> <li>SMTP Gateway</li> <li>XML Gateway</li> <li>Proxy</li> </ul>	General       T.38       G.711       Codecs       Inbound Modification Table         Options	
	G.711", at least one G.711 Codec must be in the list and all codecs that are not G.711 will be ignored.	

Select **Inbound Modification Table** tab and then select the **Add** button to add an inbound modification rule as shown in the screenshot below.

Fax	System Configuration Sites System Monitor Phone Books	
<ul> <li>SERVER3</li> <li>♀ Config Manager</li> <li>♀ Fax Manager</li> <li>♥ Driver</li> <li>● Outbound Modification Table</li> <li>♥ Channels</li> <li>♥ Channels</li> <li>♥ Rasterizer</li> <li>♀ Fax Folder Mediation</li> <li>● Fax Archive</li> <li>● Fax Archive</li> <li>● CoConfig</li> <li>♥ Fault Tolerance</li> <li>● SMTP Gateway</li> <li>● XML Gateway</li> <li>● Proxy</li> </ul>	General       T.38       G.711       Codecs       Inbound Modification Table         Settings	

KP; Reviewed: SPOC 4/1/2020

Select **Dial Plan** tab, click on **Add** button to add a pattern for outbound fax call and properly position it in the list. The dial pattern uses the start \* for any dialed number that is sent through the peer with IPO.

Fax		System Co	nfiguration	Sites	System	Monitor	Phone Books			
<ul> <li>✓ SERVER3</li> <li>○ () Config Manager</li> <li>○ () Fax Manager</li> </ul>	Driver Prope	rties								
Oriver	Options	FolP S	P SIP Secur	ity H.323	Dial Plan	Peer List	Network Capture			
Channels     Rasterizer	-Dial Plan									
<ul> <li>C Fax Folder Mediation</li> <li>C Fax Archive</li> </ul>	Dial P	lan Number Patte	m Deere	Dattern Deer	rintion	Add				
		*	10.33.1.110./10	10.33.1.110 (ID Office)		anpuon	Remove			
• 😨 Fault Tolerance			10.33.1.12 (See	sion Manager)	Avaya IP Onice		Move Up			
SMTP Gateway			ROIDDIRINE (DOI	and the starting of y			Move Down			
o 🚾 XML Gateway							Properties			
o 🧔 Proxy										

# 7. Verification Steps

The following steps may be used to verify the configuration:

From the Avaya IP Office R11 Manager screen shown in Section 5.1, select File  $\rightarrow$  Advanced  $\rightarrow$  System Status to launch the System Status application, and log in using the appropriate credentials.

The **IP Office System Status** screen is displayed. Expand **Trunks** in the left pane and select the SIP line from **Section 5.4**, in this case the SIP trunk line number is "8".

Verify that the **SIP Trunk Summary** screen shows an active channel with a **Current State** of *"Idle"*, and that the sender fax number is displayed in the **Caller ID or Dialed Digits** field.

AVAYA	i.					I	P Of	ice S	ystem S	tatus						
lp Snapshot LogOf	f Exit Abo	out														
System 🍓 Alarms (19)	Status	Utilization Su	ummary	Alarms												
Extensions (7) Trunks (6)								SIP Trun	k Summary							
Line: 1	Line Servi	ce State:		In Ser	vice											
Line: 2	Peer Dom	ain Name:		10.33	.1.60											
Line: 7	Resolved	Address:		10.33	.1.60											
Line: 8	Line Numb	er:		8												
Line: 10	Number of	f Administere	ed Channels	: 10												
Active Calls	Number of	f Channels ir	Use:	0												
tesources	Administer	red Compres	sion:	G711	Mu, G729 A											
/oicemail	Enable Fa	ststart:		Off												
PNetworking	Silence Su	ppression:		Off												
ocations	Media Stream: RTP															
	Layer 4 Protocol: UDP															
	SIP Trunk	Channel Lice	enses:	512		107.										
	SIP Trunk	Channel Lice	enses in Use	: 0		) /0										
	SIP Device	e Features:		UPDA	TE (Incoming	and Out	going)									
	Channel	URI Call	Current	Time in	Remote	Codec	Connect	. Caller ID	Other Party on	Direction	Round Trip	Receive	Receive	Transmit	Transmit	l
	Number	G Ref	State	State	Media Add			or Diale	Call	of Call	Delay	Jitter	Packet	Jitter	Packet	
	1		Idle	9 days 1											-	-
	3		Idle	9 days 1												-
	4		Idle	9 days 1												
	5		Idle	9 days 2												
	6		Idle	9 days 2												
	7		Idle	9 days 2												-
	0 0		Idle	9 days 2												
			Talla	0 4 0				-				-	-			
						_								-		_
	Trace	Trace Al	Pau	se Pin	g <u>C</u> all	Details	Grac	eful <u>S</u> hutdo	vn Force O	ut of Service	Print	5	ave As			

Verify that fax calls can be placed to/from the XM Fax server from both local and remote sites. The screenshot below shows **System Monitor** of **Outbound History** of fax calls in the XM Fax.

Control History     Control History     Control History     Control History	<b>M</b> ×	System Configuration	Sites	System Monitor d ✓ Search Show	Phone Books
ି 🛃 Outgoing Queue	Resubmit View Fax	Properties			
Services Status	1 - 50 of 64 🛯 🔹 🕨				50 per pag
• W Log Files	Completed/	Failed Time 🔻	User ID	Modified Destinatio	n Error Code
Reports	Feb 12 2020	07:56 PM	<u>user1</u>	<u>4410</u>	<u>0</u>
	Feb 12 2020	07:52 PM	<u>user1</u>	<u>914237610540</u>	<u>0</u>
	Feb 12 2020	10:36 AM	<u>user1</u>	<u>14237610540</u>	<u>0</u>
	Feb 12 2020	10:23 AM	<u>user1</u>	814237610540	<u>0</u>
	Feb 12 2020	09:26 AM	<u>user1</u>	<u>4410</u>	<u>0</u>
	Feb 11 2020	10:20 AM	<u>user1</u>	<u>814237610540</u>	<u>0</u>
	Feb 11 2020	09:45 AM	<u>user1</u>	<u>4410</u>	<u>0</u>
	Feb 8 2020 (	08:39 PM	<u>user1</u>	<u>4410</u>	<u>0</u>
	Feb 8 2020 (	08:28 PM	<u>user1</u>	<u>4410</u>	<u>0</u>
	Feb 8 2020 1	LO:43 AM	user1	<u>4410</u>	<u>0</u>
	Feb 8 2020 (	08:49 AM	<u>user1</u>	<u>411</u>	<u>0</u>
	Feb 8 2020 (	07:58 AM	<u>user1</u>	3315	<u>0</u>
	Eeb 7 2020 (	)9:57 AM	<u>user1</u>	<u>411</u>	<u>0</u>
	Feb 7 2020 (	08:14 AM	<u>user1</u>	<u>5300</u>	<u>0</u>
	Feb 7 2020 (	08:01 AM	user1	5300	<u>0</u>
	Feb 7 2020 (	)7:44 AM	<u>user1</u>	<u>5300</u>	<u>0</u>
	Eeb 6 2020 1	I ∩•39 ΔM Copyright © 2019 - XMedius Solutions In	user1 c XM Fax 9.0.0.51	3315 8	0

# 8. Conclusion

These Application Notes describe the procedures required to configure XMedius XM Fax Software to interoperate with Avaya IP Office using SIP trunks. Please refer to **Section 2.2** for any exceptions or observations.

## 9. Additional References

This section references the documentation relevant to these Application Notes. The following and additional Avaya product documentation is available at <u>http://support.avaya.com</u>.

- [1] Avaya IP Office Platform Solution Description, Release 11.0, May 2019.
- [2] Avaya IP Office Platform Feature Description, Release 11.0, May 2018.
- [3] *IP Office Platform 11.0 Deploying Avaya IP Office Essential Edition*, Document Number 15-601042, Issue 33g, 20 May 2018.
- [4] Administering Avaya IP Office Platform with Manager, Release 11.0, May 2018.
- [5] *IP Office Platform 10.1 Using Avaya IP Office Platform System Status*, Document 15-601758, Issue 13a, 05 April, 2018.
- [6] *IP Office Platform 11.0 Using IP Office System Monitor*, Document 15-601019, Issue 09b, 10 may, 2018.

XMedius XM Fax document in its most recent version may be found at <u>https://support.xmediusfax.com</u> (Sign In required)

- [1] XM Fax Installation and Maintenance Guide
- [2] XM Fax Administrator Guide Web (Web interface)
- [3] XM Fax Administrator Guide Windows (MMC Snap-In)
- [4] XM Fax User Guide

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