



FIBERME Communications LLC.

Configuring FCM630A with FAG410X

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OVERVIEW

This document describes basic configuration to interconnect FCM630A series and FAG410X. In this document, we are using FAG4108 as an example. The following methodology can be used for the FAG4104 as well. This is typically applied to the scenario where users would like to add a FAG410X not only as a remote extension but also as an external PSTN trunk.

There are two ways to set up the FCM630A series IP PBX with the FAG410X.

- **Method 1:** Configure FAG410X as a SIP Peer Trunk.
- **Method 2:** Register FAG410X on the FCM630A directly as an extension.



Warning:

When using the IVR in FCM630A series, please be aware that if "Dial Trunk" option is turned on in IVR settings, the call into the IVR will be able to dial outbound call using FCM630A's trunk. The IVR's permission level will be used when making outbound calls in this case. Please select proper permission level for the IVR to control the outbound calls allowed via "Dial Trunk".



CONNECT FCM630A TO FAG410X USING PEER SIP TRUNK

Create IVR On FCM630A

On the FCM630A web GUI, create an IVR extension under **Call Features->IVR**.

In IVR settings, if "Dial Other Extensions" is enabled, the calls dialing into the FCM630A IVR will be able to reach the internal extensions registered to the FCM630A. Also, you can assign the "Key Pressing Event" to different destinations.

Edit IVR: Welcome

Basic Settings Key Pressing Events

* Name:	FAG410X_IVR	
* Extension:	7000	
Dial Trunk:	<input type="checkbox"/>	
Auto Record:	<input type="checkbox"/>	
Dial Other Extensions:	<input type="checkbox"/> All <input checked="" type="checkbox"/> Extension <input type="checkbox"/> Multimedia Meeting <input type="checkbox"/> Call Queue <input type="checkbox"/> Ring Group <input type="checkbox"/> Paging/Intercom Groups <input type="checkbox"/> Voicemail Groups <input type="checkbox"/> Fax Extension <input type="checkbox"/> Dial By Name	
* IVR Black/Whitelist:	Disable	
Replace Display Name:	<input type="checkbox"/>	
Return to IVR Menu:	<input type="checkbox"/>	
Alert-info:	None	
* Prompt:	welcome	Upload Audio File
	Add Prompt	+
* Digit Timeout (s):	3	
* Response Timeout:	10	
* Response Timeout:	ivr-create-timeout	Upload Audio File
Prompt:		
* Invalid Input Prompt:	invalid	Upload Audio File
* Response Timeout:	3	
Prompt Repeats:		
* Invalid Input Prompt:	3	
Repeats:		
Language:	Default	

Figure 1: Create IVR 7000 on the FCM630A



Create Peer SIP TRUNK On FCM630A

On the FCM630A web GUI, create a peer SIP trunk under **Extension/Trunk ->VOIP Trunks**. In this example, the FAG410X IP address is 192.168.99.239.

Create New SIP Trunk

If the host is not a numeric IP address, but the port number is present in the URI, the FCM performs an A or AAAA record lookup of the domain name. If a domain is configured without a port number, the FCM will do an SRV record lookup.

Type:	Peer SIP Trunk
* Provider Name:	FAG410X
* Host Name:	192.168.99.239
Transport:	UDP
Keep Original CID:	<input type="checkbox"/>
Keep Trunk CID:	<input type="checkbox"/>
NAT:	<input type="checkbox"/>
Disable This Trunk:	<input type="checkbox"/>
TEL URI:	Disabled
CallerID Number:	
CallerID Name:	

Cancel Save

Figure 2: Create Peer SIP Trunk on the FCM630A

Configure Outbound Rule on FCM630A

On the FCM630A web GUI, go to **Extension/Trunk ->Outbound Routes** to create a new outbound rule. This would allow the extension on the FCM630A to reach numbers in PSTN network via the peer SIP trunk we just configured.



Create New Outbound Rule

General

* Outbound Rule Name:	FAG410X_Outbound	Disable This Route:	<input type="checkbox"/>
* Pattern:	_91XXXXXXXXXX	Privilege Level:	Internal
PIN Groups:	None	PIN Groups with Privilege Level:	
Password:		<input type="checkbox"/>	
Local Country Code:		Auto Record:	

Enable Source Caller ID Whitelist

Enable Source Caller ID Whitelist:	<input type="checkbox"/>	Outbound Route CID:	
------------------------------------	--------------------------	---------------------	--

Call Duration Limit

Call Duration Limit:	<input type="checkbox"/>
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Main Trunk

* Trunk:	SIPTrunks -- FAG410X
Strip:	1
Prepend:	

Figure 3: Configure Outbound Rule on the FCM630A

In this example pattern "91XXXXXXXXXX", 9 is the first dialing digit and it will be stripped off when the call goes out.

Configure Inbound Rule on FCM630A

On the FCM630A web GUI, go to **Extension/Trunk ->Inbound Rules** to create a new inbound rule. In this example, we create the DID as **20000**, which will be used in the FAG410X call forward setting.



Create New Inbound Rule

* Trunks:	SIPTrunks -- FAG410X	Inbound Route Name:	FAG410X_Inbound
* Pattern:	20000	CallerID Pattern:	
Disable This Route:	<input type="checkbox"/>	Seamless Transfer Whitelist:	
Alert-Info:	None	Ringback Tone:	None
Fax Detection:	<input type="checkbox"/>	Auto Record:	<input type="checkbox"/>
Block Collect Calls:	<input type="checkbox"/>	Prepend Trunk Name:	<input type="checkbox"/>
Set CallerID Info:	<input type="checkbox"/>	Enable Route-Level Inbound Mode:	<input type="checkbox"/>
Inbound Multiple Mode:	<input type="checkbox"/>		
Default Mode Mode 1			
* Default Destination:	IVR	FAG410X_IVR	
Time Condition			
<input type="button" value="Add"/>			

Figure 4: Configure Inbound Rule on FCM630A

The default destination is configured to IVR. Ensure to select the proper extension for the IVR.

Connect FAG410X with FCM630A

1. On the FAG410X web GUI, go to the **VoIP->SIP** page and enter the IP address of the FCM630A that you are peering with.

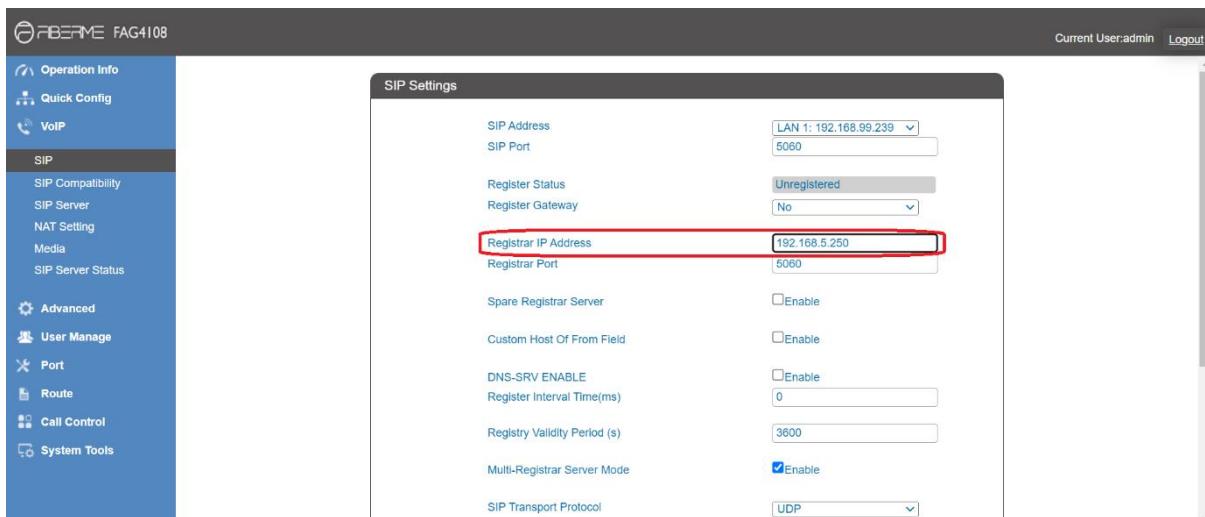


Figure 5: Connect FAG410X with FCM630A: Registrar IP Address

2. Please make sure the **Register Gateway** option under **VoIP-> SIP** is set to **No**. In the following example, FCM630A has IP address 192.168.5.250.



SIP Settings

SIP Address	LAN 1: 192.168.99.239
SIP Port	5060
Register Status	Unregistered
Register Gateway	No
Registrar IP Address	192.168.5.250
Registrar Port	5060
Spare Registrar Server	<input type="checkbox"/> Enable
Custom Host Of From Field	<input type="checkbox"/> Enable
DNS-SRV ENABLE	<input type="checkbox"/> Enable
Register Interval Time(ms)	0
Registry Validity Period (s)	3600
Multi-Registrar Server Mode	<input checked="" type="checkbox"/> Enable

Figure 6: Connect FAG410X with FCM630A: Register Gateway

Configure FXO Port on FAG410X

1. Connect the PSTN line to the FAG410X FXO port.
2. On the FAG410X web GUI, go to the **Port->FXO** page and press Modify on the FXO port you will use.

FXO Settings

Port	Type	SIP Account	Display Name	Authentication Username	Forbid Outgoing Call	Caller ID Detection	Reg Status	Echo Canceller	Server Index	Polarity Reversal Detection	Input Gain	Output Gain	Modify
1	FXO	8001	---	---	Disable	Enable	Unregistered	Enable	---	Disable	0	0	
2	FXO	8002	---	---	Disable	Enable	Unregistered	Enable	---	Disable	0	0	
3	FXO	8003	---	---	Disable	Enable	Unregistered	Enable	---	Disable	0	0	
4	FXO	8004	---	---	Disable	Enable	Unregistered	Enable	---	Disable	0	0	
5	FXO	8005	---	---	Disable	Enable	Unregistered	Enable	---	Disable	0	0	
6	FXO	8006	---	---	Disable	Enable	Unregistered	Enable	---	Disable	0	0	
7	FXO	8007	---	---	Disable	Enable	Unregistered	Enable	---	Disable	0	0	
8	FXO	8008	---	---	Disable	Enable	Unregistered	Enable	---	Disable	0	0	

8 Items Total 16 Items/Page 1/1 First Previous Next Last Go to Page 1 Pages Total Batch Modify

Figure 7: Configure FXO Port on FAG410X: Modify Port

3. Enter the SIP Account Number **20000** to be matched with the DID number we configured in FCM630A Inbound Route.

FXO-Modify

Port	1
Type	FXO
Register Port	No
SIP Account	20000
Display Name	
Authentication Username	
Display Name preferred	<input type="checkbox"/> Enable
Server Index	1:192.168.1.20
Input Gain (dB)	0
Output Gain (dB)	0
Echo Canceller	<input checked="" type="checkbox"/> Enable
Forbid Outgoing Call	<input type="checkbox"/> Enable
Caller ID Detection	<input checked="" type="checkbox"/> Enable
Polarity Reversal Detection	<input type="checkbox"/> Enable
Whitelist of FXO Out Calls	

Figure 8: Configure FXO Port on FAG410X: Port Configurations



Create Port Group on FAG410X

1. On the FAG410X web GUI, go to the **Port->Port Group** page and press “Add New”.

The screenshot shows the 'Port Group Settings' page of the FAG410X web interface. The left sidebar has a 'Port Group' section selected. The main area displays a table with one row ('default') and several configuration options below it. A red box highlights the 'Add New' button at the bottom right of the table area.

Figure 9: Create Port Group on FAG410X: Add New

2. Select the FXO port you will use.

The screenshot shows the 'Port Group-Modify' page. The left sidebar has a 'Port' section selected. The main area contains form fields for 'Index' (set to 1), 'Description' (set to 'default'), 'Register Port' (set to 'No'), and dropdowns for 'Authentication Mode' (set to 'Do Not Register') and 'Port Select Mode' (set to 'Increase'). Below these are sections for 'Port Reused by Multiple Groups' (set to 'NO') and a list of ports (checkboxes for Port 1(FXO) through Port 8(FXO)). At the bottom are 'Modify', 'Reset', and 'Cancel' buttons.

Figure 10: Create Port Group on FAG410X: Group Configurations



REGISTER FAG410X ON FCM630A AS AN EXTENSION

Create SIP Extension on FCM630A

To manually create new SIP user, go to FCM630A web GUI-> Extension/Trunk->Extensions. Click on "Add" and a new dialog window will show for users to fill in the extension information.

Extension: 1002

Basic Settings Media Features Specific Time Follow Me Cancel Save

General

* Extension: 1002 CallerID Number: 1002
* Permission: Internal * SIP/IAX Password: ****
AuthID: Voicemail: Local Voicemail
* Voicemail Password: ***** Skip Voicemail Password:
Verification:
Send Voicemail to Email: Default Keep Voicemail after: Default
Emailing:
Enable Keep-alive: * Keep-alive Frequency: 60
Disable This Extension: Enable SCA:
Emergency Calls CID:

User Settings

First Name: Last Name:
Email Address: * User Password: *****
* Language: Default * Concurrent Registrations: 1
Mobile Phone Number:

Figure 11: Create SIP Extension on FCM630A

Configure FAG410X SIP Account as a Registered Extension on FCM630A

Under FAG410X **web GUI->VoIP->SIP**, please set Register Gateway to Yes, then enter SIP Extension information created earlier in the FCM630A. In this example, extension 1002 is used in order to register FAG410X as an extension user on FCM630A.

Operation Info Quick Config VolP SIP SIP Compatibility SIP Server NAT Setting Media SIP Server Status Advanced User Manage Port Route Call Control System Tools

Current User:admin Logout

SIP Settings

SIP Address: LAN 1: 192.168.99.239 SIP Port: 5060
Register Status: Registered Register Gateway: Yes
SIP Account: 1002 Password: *****
Registrar IP Address: 192.168.5.250 Registrar Port: 5060
Spare Registrar Server: Enable
Custom Host Of From Field: Enable
DNS-SRV ENABLE: Enable
Register Interval Time(ms): 0

Figure 12: FAG410X SIP Account Settings



Under FAG410X web GUI, **VoIP->SIP**, please fill in FCM630A information as explained in method 1.

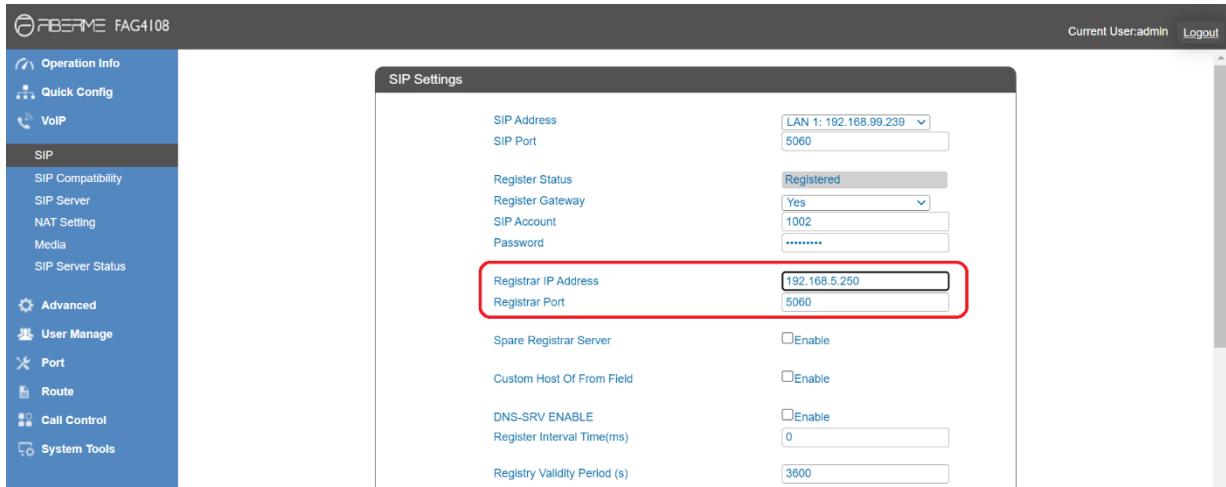


Figure 13: FAG410X Registrar IP Address

We can check FCM630A SIP Extension Status to see if FAG410X has been successfully registered as an extension device. The green icon indicates that FAG410X is registered on FCM630A.

Extensions							
		Edit		Delete		Reset	
		Edit All SIP		More		Extension Number or Name	
□	STATUS	PRESENCE S...	EXTENSION	NAME	MESSAGE	TYPE	IP AND PORT
<input type="checkbox"/>	Idle	Available	1002	0/2/0	SIP(WebRTC)	192.168.99.239:5060 (Gat...)	

Figure 14: FCM630A - SIP Extension Status

Now FAG410X is registered at FCM630A as an extension device. Please refer to method 1 in the previous section to configure FXO Port settings on FAG410X.



FAG410X CALL Routing

Configure IP to Tel on FAG410X

1. On the FAG410X web GUI, go to the **Route-> IP→TEL** page, Press “Add New”.

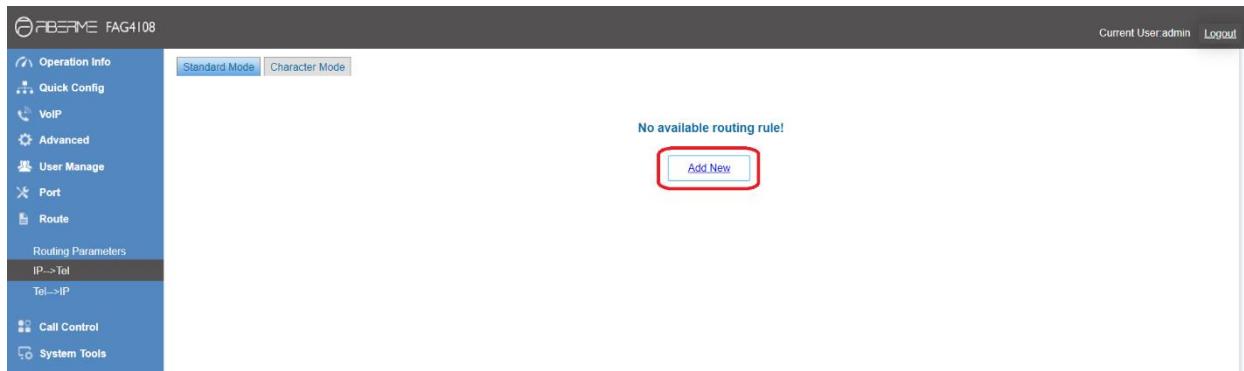


Figure 15: FAG410X – IP to TEL

2. Enter the FCM630A IP address in “Source IP” and select the Port Group you will use from “Call Destination”.

This is a configuration dialog titled "IP->Tel Routing Rule". It contains the following fields:

- Index: 63
- Description: default
- Source IP: 192.168.5.250 (highlighted with a red box)
- We suggest you input Source IP here.
- CallerID Prefix: *
- CalleeID Prefix: *
- Route by Number
- Enable
- Call Destination: 1 (highlighted with a red box)

At the bottom are "Save" and "Close" buttons.

Figure 16: IP to TEL Settings



Configure TEL to IP on FAG410X

1. On the FAG410X web GUI, go to the **Route-> TEL→IP** page, Press “Add New”.

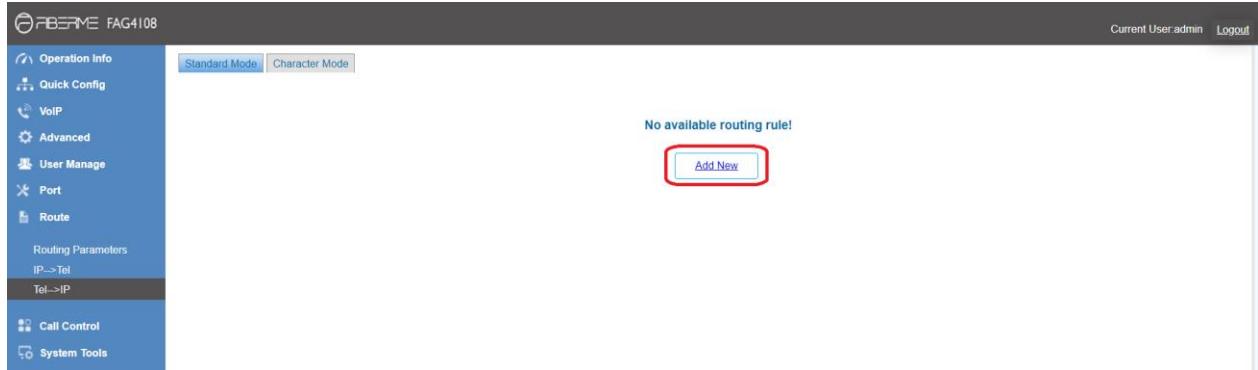


Figure 17: FAG410X – TEL to IP

2. Select the Port Group you will use from “Source Port Group” and enter the FCM630A IP address in “Destination Address”.

The dialog box is titled 'Tel->IP Routing Rule'. It contains the following fields:

- Index: 63
- Description: default
- Source Port Group: 1 (highlighted with a red box)
- CallerID Prefix: *
- CalleeID Prefix: *
- Destination Address: 192.168.5.250 (highlighted with a red box)
- Destination Port: 5060

At the bottom are 'Save' and 'Close' buttons.

Figure 18: TEL to IP Settings

