Cluana Linksys Cisco FXO adapter SPA 3102 Configuration guide

Cable connection

Step 1: connect your phone cable to the "PHONE" port. Leave the "LINE" port disconnected.



Step 2: connect your LAN cable to the "INTERNET" port. We assume a DHCP server is already running on your LAN. If not, after step 3 jump to "Static IP configuration"



Step 3: power up the adapter



IP address detection and web management

On your phone

- press **** (star key 4 time)
- press 110# and take note of the annouced IP address. Let's call it "ip_address"
- press 7932# 1# 1 (it's needed to enable web management on the internet port)
- open a web browser to the URL: http://ip_address/admin/advanced

Adapter configuration

Let's assume the followings:

Your PBX company domain is **mycompany.com** (provided by Cluana). Your PBX hostname is **myserver.cluana.com** (provided by Cluana) Your analog adapter User ID / Auth ID is **fxo_1** (your choice) Your analog adapter password is : **password** (your choice) Your phone line PSTN number is **YOUR_NUMBER** (0733784374 in the picture below)

Open SIP tab

RTP Param RTF	eters section P Packet Size:	0.020		
Router	Voice			
Info System SIP	Provisioning Regio	al Line 1 PSTN Line User 1 PSTN User	<u>User Login</u>	basic advanced
SIP Parameters				
May Essuardi	70	CUT	c	
пу васкар кост		Kery Key Koe.		
RTP Parameters				
RTP Port Min:	16384	RTP Port Max:	16482	
RTP Packet Size:	0.020	Max RTP ICMP Err:	0	

No UDP Checksum:

no 🗸

Open PSTN Line tab

0

no 🗸

RTCP Tx Interval:

Stats In BYE:

Proxy and Registration Proxy: Outbound Proxy:	mycompany.com:5061 myserver.cluana.com:5061
Register Expires:	Yes 120
Subscriber Information Display name: FXO_1 User ID: Password: Use auth ID: Auth ID:	fxo_1 password Yes fxo_1
Audio Configuration Preferred Codec:	G711a (G711u for USA)
Dial Plans Dial Plan 2: (S0<:Y	OUR_NUMBER>)
PSTN-To-VoIP Gateway Setup PSTN Ring Thru Line 1 PSTN CID For VoIP CI	: no D: yes

PSTN Caller Default DP:

FXO Timer Values (sec) PSTN Answer Delay: 2

PSTN Disconnect Detection

Disconnect Tone: US: 480@-30,620@-30;4(.25/.25/1+2)UK: 400@-30,400@-30; 2(3/0/1+2)France: 440@-30,440@-30; 2(0.5/0.5/1+2)Germany: 440@-30,440@-30; 2(0.5/0.5/1+2)Netherlands: 425@-30,425@-30; 2(0.5/0.5/1+2)Sweden: 425@-10; 10(0.25/0.25/1)Norway: 425@-10; 10(0.5/0.5/1)Italy: 425@-30,425@-30; 2(0.2/0.2/1+2)Spain: 425@-10; 10(0.2/0.2/1,0.2/0.2/1,0.2/0.6/1)Portugal: 425@-10; 10(0.5/0.5/1)Poland: 425@-10; 10(0.5/0.5/1)Denmark: 425@-10; 10(0.5/0.5/1)New Zealand: 400@-15; 10(0.25/0.25/1)Australia: 425@-13; 10(0.375/0.375/1)

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International Control

FXO Port Impedance: US: 600 EU (UK, Germany, Netherlands, Sweden, Norway, Italy, Spain, Portugal, Poland, and Denmark): 270+750||150nF France: 270+750||150nF Australia: 220+820||120nF New Zealand: 370+620||310nF

Press "Submit all changes"

Now you can connect your phone line cable to the "LINE" port.

Router	Voice		
Info System SIP Provision	ing Regional Line 1	PSTN Line User 1 DSTN User	User Login basic Ladvanced
Into System SIF Provision		oser i Port oser	
Line Enable:	ves V		
	yes -	CUT	
Proxy and Registration			
Proxy: Outbound Proxy:	mycompany.com:5061 myserver.cluana.com:5	061	
Use Outbound Proxy:	yes V	Use OB Proxy In Dialog:	yes 🗸
Register:	yes 🗸	Make Call Without Reg:	no 🔽
Register Expires:	120	Ans Call Without Reg:	
Proxy Fallback Intvl:	3600	Proxy Redundancy Method:	Normal
Subscriber Information			
Display Name:	FXO_1	User ID:	fxo_1
Password:	****	Use Auth ID:	yes 🗸
Auth ID:	fxo_1		
SRTP Private Kev:			
Audio Configuration Preferred Codec:	G711a	Silence Supp Enable:	no 🗙
		CUT	
Dial Plans Dial Plan 1:	(vv.)		
Dial Plan 2:	(S0<:0733784374>)		
Dial Plan 3:	(xx.)		
Dial Plan 4	(vv.)	CUT	
		CUT	
PSTN-To-VoIP Gateway Setup		POTH College Auth Matheda	
PSTN-TO-VOIP Gateway Enable: PSTN Ring Thru Line 1:	yes V	PSTN Caller Auth Method: PSTN PIN Max Retry:	a a a a a a a a a a a a a a a a a a a
PSTN CID For VoIP CID:	yes 🗸	PSTN CID Number Prefix:	
PSTN Caller Default DP:	2 🗸	Off Hook While Calling VoIP:	no 🔽
Line 1 Signal Hook Flash To PSTN	: Disabled	PSTN CID Name Prefix:	
		CUT	
FXO Timer Values (sec)	0	Vote DIV Digit Timorate	10
VoIP Answer Delay:	2	VOIP PIN Digit Timeout:	10
PSTN-To-VoIP Call Max Dur:	0	PSTN Ring Thru Delay:	1
VoIP-To-PSTN Call Max Dur:	0	PSTN Ring Thru CWT Delay:	3
VoIP DLG Refresh Intvl:	0	PSTN Ring Timeout:	5
PSTN Dialing Delay: PSTN Hook Flash Len:	25	PSTN Dial Digit Len:	.1/.1
Contrast Cont			
PSTN Disconnect Detection			
Detect CPC: Detect PSTN Long Silence	yes V	Detect VoIP Long Silence:	
PSTN Long Silence Duration:	30	VoIP Long Silence Duration:	30
PSTN Silence Threshold:	medium 💌	Min CPC Duration:	0.2
Detect Disconnect Tone:	yes 🗸		
Disconnect Tone:	425@-30,425@-30; 2(0	0.2/0.2/1+2)	
International Control			
FXO Port Impedance: Dtmf Playback Level:	270+750 150nF 🗸	Ring Frequency Min: Dtmf Playback Twist	1.3
SPA To PSTN Gain:	0	Ring Frequency Max:	90
PSTN To SPA Gain:	0	Ring Validation Time:	256 ms 🗸
Tip/Ring Voltage Adjust:	3.5 V 💌	Ring Indication Delay:	512 ms 💌
Operational Loop Current Min:	10 mA V	Ring Timeout:	640 ms V
Current Limiting Enable:		Ringer Impedance:	High (Normal)
Line-In-Use Voltage:	30		
	Undo All Cha	anges Submit All Changes	

Firmware Upgrade

Configuration parameters above were tested with firmware release 5.2.13. If your adapter is using an older release (check Voice menu, Info->Product Information->Software Version) maybe a software upgrade is needed.

- Download the latest firmware from the adapter manufacturer support site
- Extract the Zip file, and then run the executable file to upgrade the firmware.
- When the Firmware Upgrade Warning window appears, click Continue.
- In the next window that appears, enter the IP address of the ATA device, and then click OK.
- In the Confirm Upgrade window, verify that the correct device information and product number appear. Then click Upgrade.
- A progress message appears while the upgrade is in progress.
- The success window appears when the upgrade is completed. The device reboots. Click OK to close the confirmation message.
- To verify the upgrade, point the web browser to the IP address of the ATA device

Static IP configuration

We recommend to use a DHCP server on your Local Area Network in order to allow your adapter to automatically configure its network parameters. Should you need to assign a static IP address, please follow the Quick Install guide provided with your adapter. Remember that you need to find a free IP address compatible with your network and assign via IVR commands such IP address and its netmask (eg. 192.168.1.100 and 255.255.255.0). Such parameters should be enough to access the device via web browser and allow you to configure any other value.

Factory Reset

You can restore factory default settings with the following steps:

on your phone

- press **** (star key 4 time)
- press 73738#
- press 1
- hangup