PAP2 Phone Adapter

Installation Guide

OUTBOUND CALLING

Version 1.0

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1 Introduction

This document is an installation/configuration guide for the Linksys PAP2 phone adapter running firmware version 2.0.10 (LSc). It describes how to configure the SIP parameters, codec settings and analog coefficients for outbound calling. It also includes steps for provisioning the device with the user account and PIN. Once you complete these steps, you will be ready to begin placing calls.

2 Adapter Description

The sections below briefly describe the PAP2 device.

2.1 PAP2 Overview

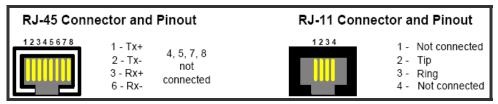
The PAP2 is a 2-port FXS phone adapter, which allows you to connect two (2) analog phones.



PAP2 2-Port FXS Phone Adapter

2.2 Connectors and Pinouts

The image below displays the Ethernet (RJ-45) and analog (RJ-11) line pin-outs.



Ethernet (RJ-45) and Analog (RJ-11) Line Pin-Outs

2.3 Accessing the PAP2

- 1. Connect an analog phone to the RJ-45 connector on the PAP2.
- 2. Using the telephone keypad, press the star key (*) four (4) times: *******. *You will be prompted to enter a code.*
- 3. Enter the appropriate code, followed by the pound (#) key. Refer to the **PAP2 Menu Options** table on the following page for codes.

	PAP2 MENU OPTIONS						
Action	Code	User Input	Behavior Notes				
Enter IVR Menu	***	None	Do not press any other keys until you hear, "Configuration menu. Please enter option followed by the pound (#) key, or hang up to exit."				
Check DHCP	100	None	The system announces if DHCP is enabled or disabled.				
Set DHCP	101	Enter 1 to enable. Enter 0 to disable.	Enter the option, followed by the pound (#) key.				
Check IP address	110	None					
Set IP Address	111	Enter value using the numbers on the telephone keypad. Use the star (*) key when entering a decimal point.	Enter the value, followed by the pound (#) key. DHCP must be "Disabled," otherwise you will hear the announcement, "Invalid Option" if you try to set this value.				
Check Net Mask	120	None					
Set Net Mask	121	Enter value using the numbers on the telephone keypad. Use the star (*) key when entering a decimal point.	Enter the value, followed by the pound (#) key. DHCP must be "Disabled," otherwise you will hear the announcement, "Invalid Option" if you try to set this value.				
Check Gateway	130	None					
Set Gateway	131	Enter the value using the numbers on the telephone keypad. Use the star (*) key when entering a decimal point.	Enter the value, followed by the pound (#) key. DHCP must be "Disabled," otherwise you will hear the announcement, "Invalid Option" if you try to set this value.				
Check MAC Address	140	None	You will hear the phone adapter's MAC address.				
Check Firmware Version	150	None	You will hear the phone adapter's firmware version.				
Manual Reboot	7326668	None	After you hear, "Option successful," hang up the phone. The phone adapter will automatically reboot.				

PAP2 Menu Options

3. Provisioning

3.1 General Configuration for Outbound Calling

To configure the PAP2 for outbound calling:

- Connect to the PAP2 by entering the device's IP address in the Web browser's Address bar. For instructions on obtaining the IP address, refer to section 2.3 Accessing the PAP2 on page 2 in this Guide.
- 2. Click on the **Admin Login** link in the upper right corner of the page, and then click the **advanced** configuration link.



PAP2 Web Interface – Main Window

3. Click the **System** tab, and in the *Internet Connection Type* section, click the **DHCP** drop-down menu, and select **yes** or **no**.

		Phone Adapter with	2 Ports for Voice Ove	er-IP	PAP
Voice	Info System S	IP Provisioning Reg	ional Line 1 Line 2	User 1 User 2	
	and the second se	Advanced View	(switch to basic view)		Üsi
tem Configurat	ion				
	Restricted Access Domeins:				
	Enable Web Server:	yes M	Web Server Port:	80	
	Enable Web Admin Access:	yes M	Admin Passwot		
	User Password:			S	
t Connection Ty	/pe				
	DHCP.	yes M			
	Static IP:		NetMask:		
	Gateway:				
Optional Netwo Configurat					
	HostName:	pap2-test	Domain:		
	Primary DNS:		Secondary DNS:		
	DNS Server Order:	Manual 💌	DNS Query Mode:	Parallel 💌	
	Syslog Server:		Debug Server:	216.53.122.54	
	Debug Level	3 💌	Primary NTP Server:		
	Secondary NTP Server:				

System Tab

4. If you set DHCP to no, enter the Static IP, NetMask, and Gateway information.

If you set DHCP to yes, continue to step 5.

5. Click the **Submit All Changes** button at the bottom of the page to save the settings.

3.2 Line-Specific Configuration

1. Click the **Line1** or **Line2** tab at the top of the page, and in the *NAT Settings* section, change the following setting:

Field	Value
NAT Keep Alive Enable	no

2. In the *Proxy and Registration* section (see the screen on the following page), change the following settings:

Field	Value
Proxy	byod1.net2phone.com
Use Outbound Proxy	no
Use OB Proxy In Dialog	no
Make Call Without Reg	yes
Ans Call Without Reg	yes
DNS SRV Auto Prefix	no

3. In the *Subscriber Information* section (as shown in the Line1/2 Window screen on the following page), enter a **Display Name**. This value can be any alphanumeric string. Enter the following values for outbound only service:

Service Type	Field	Display Name Value	
OUTBOUND SERVICE ONLY	Display Name (Line 1)	NO_DID1	
SOTBOOND SERVICE ONLY	Display Name (Line 2)	NO_DID2	

4. Enter the User ID (account) and Password (PIN).

|--|

5. Click the Use Auth ID drop-down menu, and click \mathbf{no} .

LINKSYS A Division of Cisco Systems, Inc.	0				Firmware W	ersion: 2.0.10(LSc)
		Phone Adapter with 2 P	orts for Voice-Ove	er-IP		PAP2
Voice	Info System SI	Provisioning Regiona Advanced View (swit		User 1 User 2		<u>User Login</u>
Streaming Audio Server (SAS)	Line Enable:	yes 🗸				
	SAS Enable: SAS Inbound RTP Sink:	no 💌	SAS DLG Refresh Intvl:	30		
NAT Settings	NAT Mapping Enable: NAT Keep Alive Msg:	no 💌 \$NOTIFY	NAT Keep Alive Enable: NAT Keep Alive Dest	sproxy		
Network Settings SIP Settings	SIP TOS/DiffServ Value: RTP TOS/DiffServ Value:	0x68 0xb8	Network Jitter Level:	high M		
Call Feature Settings	SIP Port: EXT SIP Port: SIP Debug Option:	5060	SIP 100REL Enable: Auth Resync-Reboot	no v yes v		
Can reature Settings	Blind Attn-Xfer Enable: Xfer When Hangup Conf:	no 💙	MOH Server:			
Proxy and Registration	Proxy: (Outbound Proxy: Register: (Register Expires: Use DNS SRV: Proxy Fallback Intvl:	et net2phone.com	Use Outbound Proxy: Use OB Proxy In Dialog: Make Call Without Reg DNS SRV Auto Prefix Voice Mail Server:	no V a: yes V		
Subscriber Information	Display Name: Password: Auth ID: Mini Certificate: SRTP Private Key:	6464324964)	User ID: Use Auth ID:	3456547898		

Proxy and Registration and Subscriber Information Sections

6. Scroll to the *Audio Configuration* section (as shown in the Audio Configurations and Dial Plan Sections screen on the following page), and change the following settings:

Field	Value
Preferred Codec	G729a
Use Pref Codec Only	no
G729a Enable	yes
G723 Enable	yes
G726-16 Enable	no
G726-24 Enable	no
G726-32 Enable	no
G726-40 Enable	no
DTMF Tx Method	AVT

Supplementary Service Subscription					
	Call Waiting Serv:	yes 🗸	Block CID Serv:	yes 🗸	
	Block ANC Serv:	yes 🗸	Dist Ring Serv:	yes 🗸	
	Cfwd All Serv:	yes 🗸	Cfwd Busy Serv:	yes 🗸	
	Cfwd No Ans Serv:	yes 🗸	Cfwd Sel Serv:	ves 🗸	
	Cfwd Last Serv:	yes 🗸	Block Last Serv:	yes 🗸	
	Accept Last Serv:	yes 🗸	DND Serv:	yes 🗸	
	CID Serv:	yes 🗸	CWCID Serv:	yes 💙	
	Call Return Serv:	yes 🕶	Call Back Serv:	yes 🕶	
	Three Way Call Serv:	yes 🕶	Three Way Conf Serv:	yes 🕶	
	Attn Transfer Serv:	yes 🕶	Unattn Transfer Serv:	yes 🕶	
	MVI Serv:	yes 🔽	VMVI Serv:	yes 🕶	
	Speed Dial Serv:	yes 💌	Secure Call Serv:	yes 🔽	
	Referral Serv:	yes 💌	Feature Dial Serv:	yes 💌	
Audio Configuration		\frown			
	Preferred Codec:	G729a	Silence Supp Enable:	no 💌	
	Use Pref Codec Only	no 💌	Silence Threshold:	medium 💌	
	G729a Enable:	yes 🕶	Echo Canc Enable:	yes 🕶	
	G723 Enable:	yes 🕶	Echo Canc Adapt Enable:	yes 🕶	
	G726-16 Enable:	no 💌	Echo Supp Enable:	yes 🕶	
	G726-24 Enable:	no 💌	FAX CED Detect Enable:	yes 💙	
	G726-32 Enable:	no 💌	FAX CNG Detect Enable:	yes 🗸	
	G726-40 Enable:	no 💌	Enaple: FAX Passthru Codec:	G711u 💌	
	DTMF Tx Method:	AVT VA	FAX Codec Symmetric:	yes 🕶	
	Hook Flash Tx Method:	None 👻	FAX Passthru Method:	NSE 💌	
	Release Unused Codec:	yes 🕶	FAX Process NSE:	yes 💙	
Dial Plan					
(Dial Plan:	(*xx [3469]11 1xxx[2-9]xxxxxx <:0	11>[2-9]x. xxxxxxxxxxxxx		
	Enable IP Dialing:	no 💌			
FXS Port Polarity Configuration					
	Idle Polarity:	Forward 💌	Caller Conn Polarity:	Forward 💌	
	Callee Conn Polarity:	Forward 💌			
					CISCO SYSTEMS
		Course Continent -	Connel Rotting		ավկությո
		Save Settings	Cancel Setting	S	

Audio Configuration and Dial Plan Sections

- In the Dial Plan section, change the Dial Plan to fit the dialing pattern in your country. The dialed number must start with '1' for US calls and '011' for non-US calls. Please read the PAP2 manual to customize your dial plan.
- 8. Click the **Submit All Changes** button to save the settings.