

Linksys Phone Adapter Configuration

Router Voice

Info System SIP Provi	sioning Regional Line 1 PS7	N Line User 1 PSTN User	<u>User Login</u> <u>basic</u> advan
ine Enable:	yes ▼		
Streaming Audio Server (SA			
SAS Enable:	no ▼	SAS DLG Refresh Intvl:	30
SAS Inbound RTP Sink:			
NAT Settings			
IAT Mapping Enable:	no ▼	NAT Keep Alive Enable:	no ▼
IAT Keep Alive Msg:	\$NOTIFY	NAT Keep Alive Dest:	\$PROXY
letwork Settings			
SIP ToS/DiffServ Value:	0x68	SIP CoS Value:	3 [0-7]
RTP ToS/DiffServ Value:	0xb8	RTP CoS Value:	6 [0-7]
letwork Jitter Level:	high ▼	Jitter Buffer Adjustment:	up and down ▼
SIP Settings			
SIP Transport:	UDP ▼	SIP Port:	5060
SIP 100REL Enable:	no ▼	EXT SIP Port:	
auth Resync-Reboot:	yes ▼	SIP Proxy-Require:	
IP Remote-Party-ID:	yes ▼	SIP GUID:	no ▼
IP Debug Option:	none	▼ RTP Log Intvl:	0
estrict Source IP:	no ▼	Referor Bye Delay:	4
efer Target Bye Delay:	0	Referee Bye Delay:	0
efer-To Target Contact:	no ▼	Sticky 183:	no ▼
uth INVITE:	no ▼	Reply 182 On Call Waiting:	no ▼
Call Feature Settings			
Blind Attn-Xfer Enable:	no ▼	MOH Server:	
(fer When Hangup Conf:	yes ▼		
Proxy and Registration			
Proxy:	host_name:5160		
Outbound Proxy:	_		
Jse Outbound Proxy:	no ▼	Use OB Proxy In Dialog:	yes ▼
Register:	yes ▼	Make Call Without Reg:	yes ▼
legister Expires:	3600	Ans Call Without Reg:	yes ▼
Ise DNS SRV:	no ▼	DNS SRV Auto Prefix:	no ▼
roxy Fallback Intvl:	3600	Proxy Redundancy Method:	Normal ▼
oice Mail Server:		Mailbox Subscribe Expires:	2147483647
subscriber Information			
isplay Name:	number	User ID:	number
assword:	******	Use Auth ID:	no ▼
uth ID:	number		
1ini Certificate:			
SRTP Private Key:			
Supplementary Service Subs	scription		
Call Waiting Serv:	yes ▼	Block CID Serv:	yes ▼
Block ANC Serv:	yes ▼	Dist Ring Serv:	yes ▼
Cfwd All Serv:	yes ▼	Cfwd Busy Serv:	yes ▼
	•		•
fwd No Ans Serv:	yes ▼	Cfwd Sel Serv:	yes ▼

5.07.2019		Linksys SPA Configuration	
Accept Last Serv:	yes ▼	DND Serv:	yes ▼
CID Serv:	yes ▼	CWCID Serv:	yes ▼
Call Return Serv:	yes ▼	Call Redial Serv:	yes ▼
Call Back Serv:	yes ▼	Three Way Call Serv:	yes ▼
Three Way Conf Serv:	yes ▼	Attn Transfer Serv:	yes ▼
Unattn Transfer Serv:	yes ▼	MWI Serv:	yes ▼
VMWI Serv:	yes ▼	Speed Dial Serv:	yes ▼
Secure Call Serv:	yes ▼	Referral Serv:	yes ▼
Feature Dial Serv:	yes ▼	Service Announcement Serv:	no ▼
Audio Configuration			
Preferred Codec:	G711u ▼	Second Preferred Codec:	Unspecified ▼
Third Preferred Codec:	Unspecified ▼	Use Pref Codec Only:	no ▼
Silence Supp Enable:	no ▼	Silence Threshold:	medium ▼
G729a Enable:	yes ▼	Echo Canc Enable:	yes ▼
G723 Enable:	yes ▼	Echo Canc Adapt Enable:	yes ▼
G726-16 Enable:	yes ▼	Echo Supp Enable:	yes ▼
G726-24 Enable:	yes ▼	FAX CED Detect Enable:	yes ▼
G726-32 Enable:	yes ▼	FAX CNG Detect Enable:	yes ▼
G726-40 Enable:	yes ▼	FAX Passthru Codec:	G711u ▼
DTMF Process INFO:	yes ▼	FAX Codec Symmetric:	yes ▼
DTMF Process AVT:	yes ▼	FAX Passthru Method:	NSE ▼
DTMF Tx Method:	Auto ▼	DTMF Tx Mode:	Strict ▼
DTMF Tx Strict Hold Off Time:	40	FAX Process NSE:	yes ▼
Hook Flash Tx Method:	None ▼	FAX Disable ECAN:	no ▼
Release Unused Codec:	yes ▼	FAX Enable T38:	yes ▼
FAX T38 Redundancy:	1 ▼	FAX Tone Detect Mode:	caller or callee ▼
Symmetric RTP:	yes ▼		
Gateway Accounts			
Gateway 1:		GW1 NAT Mapping Enable:	no ▼
GW1 Auth ID:		GW1 Password:	
Gateway 2:		GW2 NAT Mapping Enable:	no ▼
GW2 Auth ID:		GW2 Password:	
Gateway 3:		GW3 NAT Mapping Enable:	no ▼
GW3 Auth ID:		GW3 Password:	
Gateway 4:		GW4 NAT Mapping Enable:	no ▼
GW4 Auth ID:		GW4 Password:	
VoIP Fallback To PSTN			
Auto PSTN Fallback:	yes ▼		
Dial Plan			
Dial Plan:	(*xx [3469]11 0 00 [2-9]xxxxxx 1xxx[2-9]xxxxxxS0 xxxxxxxxxxxxxxxxxxxxxxxxxxxxxx		
Enable IP Dialing:	no ▼	Emergency Number:	

FXS Port Polarity Configuration

Idle Polarity: Forward ▼ Caller Conn Polarity: Forward ▼

Callee Conn Polarity: Forward ▼

Undo All Changes

Submit All Changes

<u>User Login</u> <u>basic</u> | advanced

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