



## Linksys Phone Adapter Configuration

Router

**Voice**

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Line Enable:	yes		
<b>Streaming Audio Server (SAS)</b>			
SAS Enable:	no	SAS DLG Refresh Intvl:	30
SAS Inbound RTP Sink:			
<b>NAT Settings</b>			
NAT Mapping Enable:	yes	NAT Keep Alive Enable:	yes
NAT Keep Alive Msg:	\$NOTIFY	NAT Keep Alive Dest:	\$PROXY
<b>Network Settings</b>			
SIP ToS/DiffServ Value:	0x68	SIP CoS Value:	3 [0-7]
RTP ToS/DiffServ Value:	0xb8	RTP CoS Value:	6 [0-7]
Network Jitter Level:	high	Jitter Buffer Adjustment:	up and down
<b>SIP Settings</b>			
SIP Transport:	UDP	SIP Port:	5060
SIP 100REL Enable:	no	EXT SIP Port:	
Auth Resync-Reboot:	yes	SIP Proxy-Require:	
SIP Remote-Party-ID:	yes	SIP GUID:	no
SIP Debug Option:	none	RTP Log Intvl:	0
Restrict Source IP:	no	Referor Bye Delay:	4
Refer Target Bye Delay:	0	Referee Bye Delay:	0
Refer-To Target Contact:	no	Sticky 183:	no
Auth INVITE:	no	Reply 182 On Call Waiting:	no
Use Anonymous With RPID:	yes	Use Local Addr In FROM:	no
<b>Call Feature Settings</b>			
Blind Attn-Xfer Enable:	no	MOH Server:	
Xfer When Hangup Conf:	yes		
<b>Proxy and Registration</b>			
Proxy:	sip.iptel.org		
Outbound Proxy:	sip.iptel.org		
Use Outbound Proxy:	yes	Use OB Proxy In Dialog:	yes
Register:	yes	Make Call Without Reg:	no
Register Expires:	1800	Ans Call Without Reg:	no
Use DNS SRV:	yes	DNS SRV Auto Prefix:	yes
Proxy Fallback Intvl:	3600	Proxy Redundancy Method:	Normal
Voice Mail Server:		Mailbox Subscribe Expires:	2147483647
<b>Subscriber Information</b>			
Display Name:	xxxxx	User ID:	xxxxxxxxx
Password:	*****	Use Auth ID:	yes

Auth ID: xxxxxxxx

Mini Certificate:

SRTP Private Key:

**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:	yes
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 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:	G729a	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 T38 V29 Only:	no
FAX Tone Detect Mode:	caller or callee	Audio Dump Option1:	none
Audio Dump Option2:	none	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|xxxxxxxxxxxxx.)

Enable IP Dialing: yes      Emergency Number:

### FXS Port Polarity Configuration

Idle Polarity: Forward

Caller Conn Polarity: Forward

Callee Conn Polarity: Forward

Undo All Changes

Submit All Changes

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