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Linksys Phone Adapter Configuration

Router **Voice**

Info System SIP Provisioning Regional **Line 1** PSTN Line User 1 PSTN User [User Login](#) [basic](#) | [advan](#)

Line Enable: yes

Streaming Audio Server (SAS)

SAS Enable: no

SAS DLG Refresh Intvl: 30

SAS Inbound RTP Sink:

NAT Settings

NAT Mapping Enable: no

NAT Keep Alive Enable: no

NAT Keep Alive Msg: \$NOTIFY

NAT Keep Alive Dest: \$PROXY

Network Settings

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

SIP Settings

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

Call Feature Settings

Blind Attn-Xfer Enable: no

MOH Server:

Xfer When Hangup Conf: yes

Proxy and Registration

Proxy: sip.voipdiscount.com

Outbound Proxy:

Use Outbound Proxy: no

Use OB Proxy In Dialog: yes

Register: yes

Make Call Without Reg: yes

Register Expires: 3600

Ans Call Without Reg: no

Use DNS SRV: no

DNS SRV Auto Prefix: no

Proxy Fallback Intvl: 3600

Proxy Redundancy Method: Normal

Voice Mail Server:

Mailbox Subscribe Expires: 2147483647

Subscriber Information

Display Name: syntiksi

User ID: syntiksi

Password: \*\*\*\*\*

Use Auth ID: no

Auth ID:

192.168.2.2

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:	G711u	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:	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
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:	sip.voipdiscount.com	GW1 NAT Mapping Enable:	no
GW1 Auth ID:	syntiksi	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: L:15,S:10,(S0 <:0030>2xxxxxxxx | S0 <:0030>69xxxxxxxx | S0 8xxxx)  
 Enable IP Dialing: no      Emergency Number:

FXS Port Polarity Configuration

Idle Polarity:	Forward	Caller Conn Polarity:	Forward
Callee Conn Polarity:	Forward		