



Linksys Phone Adapter Configuration

Router

Voice

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Line Enable: yes ▼

NAT Settings

NAT Mapping Enable: no ▼

NAT Keep Alive Msg: \$NOTIFY

NAT Keep Alive Enable: no ▼

NAT Keep Alive Dest: \$PROXY

Network Settings

SIP ToS/DiffServ Value: 0x68

RTP ToS/DiffServ Value: 0xb8

Network Jitter Level: high ▼

SIP CoS Value: 3 [0-7]

RTP CoS Value: 6 [0-7]

Jitter Buffer Adjustment: up and down ▼

SIP Settings

SIP Transport: UDP ▼

SIP 100REL Enable: no ▼

Auth Resync-Reboot: yes ▼

SIP Remote-Party-ID: yes ▼

SIP Debug Option: none ▼

Restrict Source IP: no ▼

Refer Target Bye Delay: 0

Refer-To Target Contact: no ▼

Auth INVITE: no ▼

SIP Port: 5061

EXT SIP Port:

SIP Proxy-Require:

SIP GUID: no ▼

▼ RTP Log Intvl: 0

Referor Bye Delay: 4

Referee Bye Delay: 0

Sticky 183: no ▼

Proxy and Registration

Proxy: host_name

Outbound Proxy:

Use Outbound Proxy: no ▼

Register: yes ▼

Register Expires: 3600

Use DNS SRV: no ▼

Proxy Fallback Intvl: 3600

Use OB Proxy In Dialog: yes ▼

Make Call Without Reg: yes ▼

Ans Call Without Reg: yes ▼

DNS SRV Auto Prefix: no ▼

Proxy Redundancy Method: Normal ▼

Subscriber Information

Display Name:

Password: *****

Auth ID:

Mini Certificate:

SRTP Private Key:

User ID: username

Use Auth ID: yes ▼

Audio Configuration

Preferred Codec: G711u ▼

Use Pref Codec Only: no ▼

G729a Enable: yes ▼

G723 Enable: yes ▼

G726-16 Enable: yes ▼

G726-24 Enable: yes ▼

G726-32 Enable: yes ▼

G726-40 Enable: yes ▼

DTMF Process INFO: yes ▼

DTMF Process AVT: yes ▼

DTMF Tx Mode: Strict ▼

Release Unused Codec: yes ▼

Symmetric RTP: yes ▼

Silence Supp Enable: no ▼

Echo Canc Enable: yes ▼

Echo Canc Adapt Enable: yes ▼

Echo Supp Enable: yes ▼

FAX CED Detect Enable: yes ▼

FAX CNG Detect Enable: yes ▼

FAX Passthru Codec: G711u ▼

FAX Codec Symmetric: yes ▼

FAX Passthru Method: NSE ▼

DTMF Tx Method: Auto ▼

DTMF Tx Strict Hold Off Time: 40

FAX Process NSE: yes ▼

FAX Disable ECAN: no ▼

Dial Plans

Dial Plan 1: (xx.)
 Dial Plan 2: (S0<:did_number>)
 Dial Plan 3: (xx.)
 Dial Plan 4: (xx.)
 Dial Plan 5: (xx.)
 Dial Plan 6: (xx.)
 Dial Plan 7: (xx.)
 Dial Plan 8: (xx.)

VoIP-To-PSTN Gateway Setup

VoIP-To-PSTN Gateway Enable:	yes ▼	VoIP Caller Auth Method:	none ▼
VoIP PIN Max Retry:	3	One Stage Dialing:	yes ▼
Line 1 VoIP Caller DP:	1 ▼	VoIP Caller Default DP:	1 ▼
Line 1 Fallback DP:	none ▼		
VoIP Caller ID Pattern:			
VoIP Access List:			
VoIP Caller 1 PIN:		VoIP Caller 1 DP:	1 ▼
VoIP Caller 2 PIN:		VoIP Caller 2 DP:	1 ▼
VoIP Caller 3 PIN:		VoIP Caller 3 DP:	1 ▼
VoIP Caller 4 PIN:		VoIP Caller 4 DP:	1 ▼
VoIP Caller 5 PIN:		VoIP Caller 5 DP:	1 ▼
VoIP Caller 6 PIN:		VoIP Caller 6 DP:	1 ▼
VoIP Caller 7 PIN:		VoIP Caller 7 DP:	1 ▼
VoIP Caller 8 PIN:		VoIP Caller 8 DP:	1 ▼

VoIP Users and Passwords (HTTP Authentication)

VoIP User 1 Auth ID:	VoIP User 1 DP:	1 ▼
VoIP User 1 Password:		
VoIP User 2 Auth ID:	VoIP User 2 DP:	1 ▼
VoIP User 2 Password:		
VoIP User 3 Auth ID:	VoIP User 3 DP:	1 ▼
VoIP User 3 Password:		
VoIP User 4 Auth ID:	VoIP User 4 DP:	1 ▼
VoIP User 4 Password:		
VoIP User 5 ID Auth ID:	VoIP User 5 DP:	1 ▼
VoIP User 5 Password:		
VoIP User 6 Auth ID:	VoIP User 6 DP:	1 ▼
VoIP User 6 Password:		
VoIP User 7 Auth ID:	VoIP User 7 DP:	1 ▼
VoIP User 7 Password:		
VoIP User 8 Auth ID:	VoIP User 8 DP:	1 ▼
VoIP User 8 Password:		

PSTN-To-VoIP Gateway Setup

PSTN-To-VoIP Gateway Enable:	yes ▼	PSTN Caller Auth Method:	none ▼
PSTN Ring Thru Line 1:	no ▼	PSTN PIN Max Retry:	3
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:	
PSTN Caller ID Pattern:			
PSTN Access List:			
PSTN Caller 1 PIN:		PSTN Caller 1 DP:	1 ▼
PSTN Caller 2 PIN:		PSTN Caller 2 DP:	1 ▼
PSTN Caller 3 PIN:		PSTN Caller 3 DP:	1 ▼
PSTN Caller 4 PIN:		PSTN Caller 4 DP:	1 ▼
PSTN Caller 5 PIN:		PSTN Caller 5 DP:	1 ▼
PSTN Caller 6 PIN:		PSTN Caller 6 DP:	1 ▼
PSTN Caller 7 PIN:		PSTN Caller 7 DP:	1 ▼
PSTN Caller 8 PIN:		PSTN Caller 8 DP:	1 ▼

FXO Timer Values (sec)

VoIP Answer Delay:	0	VoIP PIN Digit Timeout:	10
PSTN Answer Delay:	16	PSTN 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:	1	PSTN Dial Digit Len:	.1/.1
PSTN Hook Flash Len:	.25		
PSTN Disconnect Detection			
Detect CPC:	yes ▼	Detect Polarity Reversal:	yes ▼
Detect PSTN Long Silence:	no ▼	Detect VoIP Long Silence:	no ▼
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:	480@-30,620@-30;4(.25/.25/1+2)		
International Control			
FXO Port Impedance:	600 ▼	Ring Frequency Min:	10
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 ▼	Ring Timeout:	640 ms ▼
On-Hook Speed:	Less than 0.5 ms ▼	Ring Threshold:	13.5-16.5 Vrms ▼
Current Limiting Enable:	no ▼	Ringer Impedance:	High (Normal) ▼
Line-In-Use Voltage:	30		

Undo All Changes

Submit All Changes