



BBPGlobal
Add voice to your internet

Linksys SPA941 VoIP Phone Configuration Instructions


Version 1.0

A. Physical Connection

1. Connect the device as illustrated



B. VoIP Settings

1. Press setup button  on the phone and press 9 to obtain Current IP address
2. Login the VoIP phone via any web browser e.g. Internet Explorer by typing in the IP address
3. Login advanced mode and Click Ext 1
4. Input the information as illustrated


Sipura Telephone Configuration

Info System SIP Provisioning Regional Phone **Ext 1** Ext 2 User
User Login basic | advanced
Personal Directory Call History

General

Line Enable:

Share Line Appearance

Share Ext: Shared User ID:

Subscription Expires:

NAT Settings

NAT Mapping Enable: NAT Keep Alive Enable:

NAT Keep Alive Msg: NAT Keep Alive Dest:

Network Settings

SIP TOS/DiffServ Value: SIP CoS Value:

RTP TOS/DiffServ Value: RTP CoS Value:

Network Jitter Level: Jitter Buffer Adjustment:

SIP Settings

SIP Port: SIP 100REL Enable:

EXT SIP Port: Auth Resync-Reboot:

SIP Proxy-Require: SIP Remote-Party-ID:

Referor Bye Delay: Refer-To Target Contact:

Referee Bye Delay: SIP Debug Option:

Refer Target Bye Delay: Sticky 183:

Call Feature Settings

Blind Attn-Xfer Enable: MOH Server:

Message Waiting: Auth Page:

Default Ring: Auth Page Realm:

Conference Bridge URL: Auth Page Password:

Mailbox ID: Voice Mail Server:

State Agent: CFWD Notify Serv:

CFWD Notifier:

Proxy and Registration			
Proxy:	sip2.bbpglobal.com	Use Outbound Proxy:	no
Outbound Proxy:		Use OB Proxy In Dialog:	yes
Register:	yes	Make Call Without Reg:	yes
Register Expires:	120	Ans Call Without Reg:	yes
Use DNS SRV:	no	DNS SRV Auto Prefix:	no
Proxy Fallback Intvl:	3600	Proxy Redundancy Method:	Normal
Subscriber Information			
Display Name:	Extension #	User ID:	Extension #
Password:	Password	Use Auth ID:	yes
Auth ID:	Extension #		
Mini Certificate:			
S RTP Private Key:			
Audio Configuration			
Preferred Codec:	G729a	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
Release Unused Codec:	yes	DTMF Process AVT:	yes
Silence Supp Enable:	no	DTMF Tx Method:	Auto
Dial Plan			
Dial Plan:	(*xx [3469]11 0 00 [2-9]xxxxxx 1xxx[2-9]xxxxxxS0 xxxxxxxxxxxxx.)		
Enable IP Dialing:	yes		

5. Click Submit All Changes