



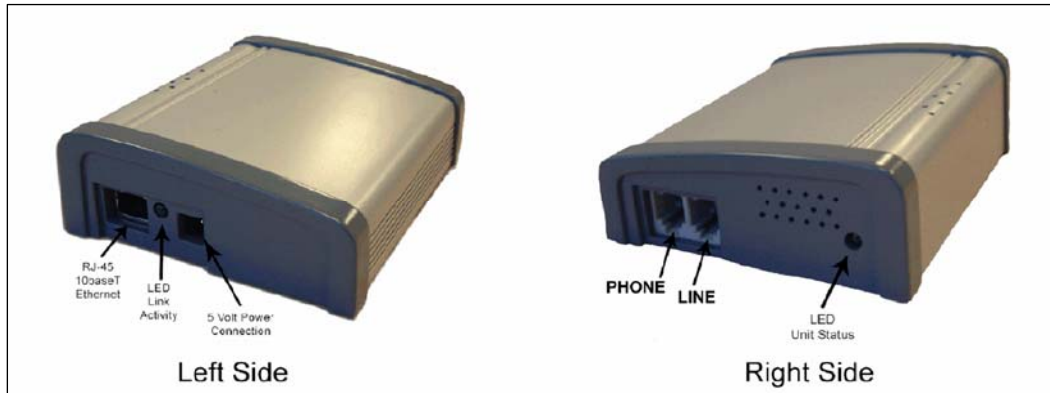
# Sipura SPA-3000 Phone Adapter Configuration Instructions

Version 1.0

## Configuration for Sipura Phone Adapter SPA-3000

### A. Physical connection

#### 1. Connect the Phone Adapter

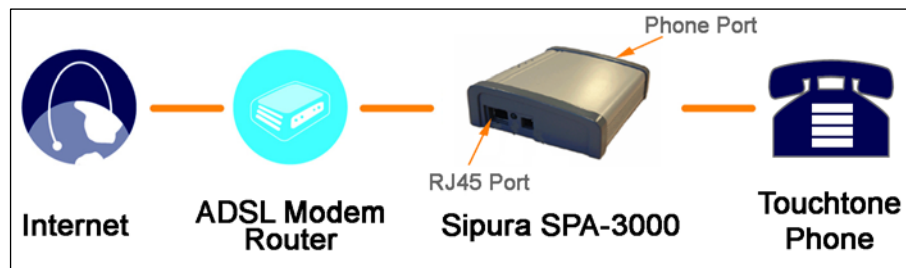


From the left side of the SPA

- Insert a standard RJ-45 Ethernet cable (included) into the LAN port
- Insert the power adapter cable into the 5V power adapter cable receptacle. Ensure that the power adapter jack is snugly attached to the SPA-3000.

From the right side of the SPA

- Insert a standard RJ-11 telephone cable into the PHONE port
- Connect the other end of the cable to an analog telephone



NB. Do not connect RJ-11 telephone cable from the SPA-3000 PHONE port to the wall jack to prevent any chance of connection to the POTS circuit switched telco network.

You may now insert the plug end of the power adapter into a live power outlet which will power up the SPA.

## B. Enter the Configuration Menu

1. Look for IP Address of the phone adapter
  - Pick up the phone and type \* \* \* \* (star key x 4)
  - Type 110#
  - The IP address would be read out
2. Type the IP address onto the address bar of Internet Explorer
3. Click “Admin Login” as illustrated



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A Division of Cisco Systems, Inc.

*Linksys Phone Adapter Configuration*

Info System User 1 PSTN User [Admin Login](#) basic advanced

**System Information**

DHCP:	Enabled	Current IP:	
Host Name:	SipuraSPA	Domain:	
Current Netmask:	255.255.255.0	Current Gateway:	192.168.0.1
Primary DNS:	202.68.78.3		
Secondary DNS:	192.189.54.17 192.189.52.33		

**Product Information**

Product Name:	SPA-3000	Serial Number:	88012DA33693
Software Version:	3.1.10(GWd)	Hardware Version:	2.0.1(5509)
MAC Address:	000E08CA40EC	Client Certificate:	Installed

**System Status**

Current Time:	5/18/2006 09:02:29	Elapsed Time:	00:15:42
Broadcast Pkts Sent:	0	Broadcast Bytes Sent:	0
Broadcast Pkts Recv:	5004	Broadcast Bytes Recv:	558858
Broadcast Pkts Dropped:	0	Broadcast Bytes Dropped:	0

### C. System Settings

1. Click on “System” Tab and enter the information as illustrated. Click “Submit All Changes” when done.



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

System Configuration  
 Enable Web Server:  User Password:

Internet Connection Type  
 DHCP:   
 Static IP:  NetMask:   
 Gateway:

Optional Network Configuration  
 HostName:  Domain:   
 Primary DNS:  Secondary DNS:   
 DNS Query Mode:  Syslog Server:   
 Debug Server:  Debug Level:

[User Login](#) basic | [advanced](#)

### D. SIP Settings

1. Click on “SIP” Tab and enter the information as illustrated. Click “Submit All Changes” when done.



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

Response Status Code Handling  
 SIT1 RSC:  SIT2 RSC:   
 SIT3 RSC:  SIT4 RSC:

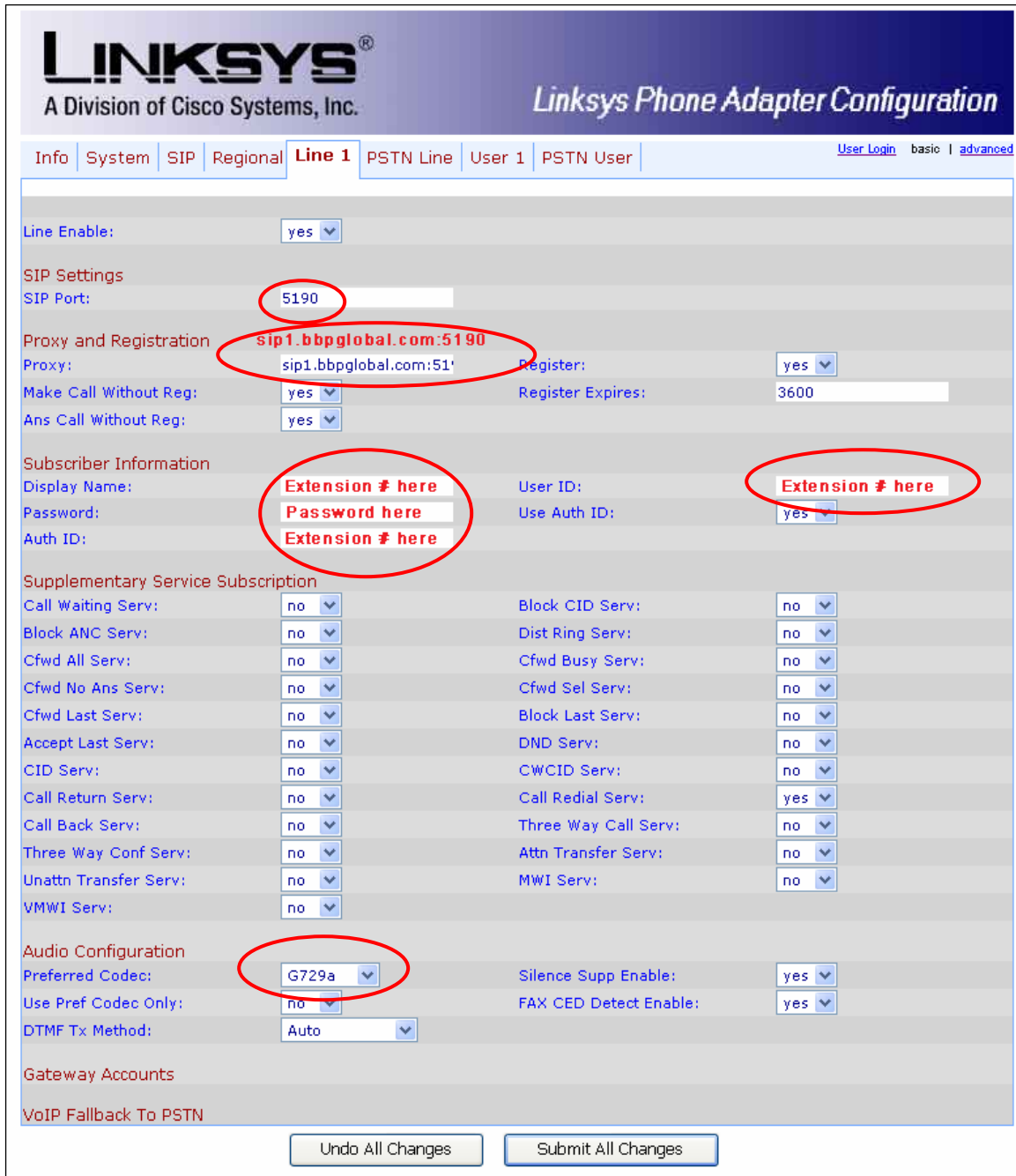
RTP Parameters  
 RTP Port Min:  RTP Port Max:

SDP Payload Types  
 NSE Dynamic Payload:  AVT Dynamic Payload:   
 G726r16 Dynamic Payload:  G726r24 Dynamic Payload:   
 G726r40 Dynamic Payload:  G729b Dynamic Payload:

[User Login](#) basic | [advanced](#)

## E. Configure Line 1 or 2

1. Click on “Line 1” Tab and enter the information as illustrated. Click “Submit All Changes” when done.



**LINKSYS**  
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*Linksys Phone Adapter Configuration*

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

Line Enable:

SIP Settings  
SIP Port:

Proxy and Registration  
Proxy:  Register:  Register Expires:

Make Call Without Reg:  Ans Call Without Reg:

Subscriber Information  
Display Name:  User ID:   
Password:  Use Auth ID:   
Auth ID:

Supplementary Service Subscription

Call Waiting Serv:	<input type="text" value="no"/>	Block CID Serv:	<input type="text" value="no"/>
Block ANC Serv:	<input type="text" value="no"/>	Dist Ring Serv:	<input type="text" value="no"/>
Cfwd All Serv:	<input type="text" value="no"/>	Cfwd Busy Serv:	<input type="text" value="no"/>
Cfwd No Ans Serv:	<input type="text" value="no"/>	Cfwd Sel Serv:	<input type="text" value="no"/>
Cfwd Last Serv:	<input type="text" value="no"/>	Block Last Serv:	<input type="text" value="no"/>
Accept Last Serv:	<input type="text" value="no"/>	DND Serv:	<input type="text" value="no"/>
CID Serv:	<input type="text" value="no"/>	CWCID Serv:	<input type="text" value="no"/>
Call Return Serv:	<input type="text" value="no"/>	Call Redial Serv:	<input type="text" value="yes"/>
Call Back Serv:	<input type="text" value="no"/>	Three Way Call Serv:	<input type="text" value="no"/>
Three Way Conf Serv:	<input type="text" value="no"/>	Attn Transfer Serv:	<input type="text" value="no"/>
Unattn Transfer Serv:	<input type="text" value="no"/>	MWI Serv:	<input type="text" value="no"/>
VMWI Serv:	<input type="text" value="no"/>		

Audio Configuration  
Preferred Codec:  Silence Supp Enable:   
Use Pref Codec Only:  FAX CED Detect Enable:   
DTMF Tx Method:

Gateway Accounts

VoIP Fallback To PSTN

NB: Port number of sip2.bbpglobal.com is 5060.