

Configuration Guide for the Sipura SPA-2000



CONNECT THE SPA-2000

Note:					
<i>Firmware 2.0.13(g) is required for service.</i>					

- 1. Using a standard telephone cable (RJ-11), connect an analog phone to the **Phone1** port on the SPA-2000.
- 2. Using a standard network cable (RJ-45), connect the LAN port on the SPA-2000 to your network connection (e.g., LAN).
- 3. Power on the SPA-2000 by plugging its power supply into an AC outlet and then plugging the power cable into the **Power** port on the SPA-2000.
- 4. When the system announces an Interactive Voice Response (IVR), enter the code that corresponds with the action you would like to perform, followed by the pound (#) key. Refer to the SPA-2000 Menu Options table below for codes.

SPA-2000 MENU OPTIONS							
ACTION	CODE	USER INPUT	NOTES				
Enter IVR Menu	****	N/A	Wait until you hear the announcement, "Sipura configuration menu. Please enter option followed by the pound (#) key, or hang up to exit."				
Check DHCP	100	N/A	You will hear the system announce the DHCP status, either enabled or disabled.				
Set DHCP	101	To enable, enter 1. To disable, enter 0.	Press the number corresponding with the desired option, followed by the pound (#) key.				
Check IP address	110	N/A					
Set IP Address	111	Use the phone's keypad to enter the IP address. Press the star (*) key to enter the "dots" in the IP address.	After entering the IP address, press the pound (#) key. Ensure that DHCP is "Disabled." If it is "Enabled," the system announces, "Invalid Option" if you try to set this value.				
Check Net Mask	120	N/A					
Set Net Mask	121	Use the phone's keypad to enter the Net Mask address. Press the star (*) key to enter the "dots" in the IP address.	After entering the Net Mask address, press the pound (#) key. Ensure that DHCP is "Disabled." If it is "Enabled," the system announces, "Invalid Option" if you try to set this value.				
Check Gateway	130	N/A					
Set Gateway	131	Use the phone's keypad to enter the gateway address. Press the star (*) key to enter the "dots" in the IP address.	After entering the gateway address, press the pound (#) key. Ensure that DHCP is "Disabled." If it is "Enabled," the system announces, "Invalid Option" if you try to set this value.				

CONFIGURE THE SPA-2000

OBTAIN THE SPA-2000'S IP ADDRESS

To obtain the SPA-2000's IP address:

- 1. Pick up the handset of the phone connected to the SPA-2000, and then, using the phone's keypad, press the star key (*) four times: ****.
- 2. When you hear an IVR prompt, press **110#**. When the system announces the SPA-2000's IP address, write it down.

CONNECT TO THE SPA-2000 WEB INTERFACE

To connect to the SPA-2000 Web interface and configure the SPA-2000:

- 1. Using a PC on the same network as the SPA-2000, launch a Web browser, and in the **Address** bar, enter the SPA-2000's IP address (obtained in the previous section).
- 2. Click the Admin Login link, located in the upper right-hand corner of the page.
- 3. Click the **switch to advanced view** link.
- 4. Click the tab for Line1 or Line2 at the top of the page, depending on which line you are configuring.
- 5. In the **NAT Settings** section, change the value of the following field:
 - NAT Keep Alive Enable no
- 6. In the **Proxy and Registration** section, change the following fields to the value indicated:
 - Proxy sip.primetalker.com
 - Use Outbound Proxy no
 - Use OB Proxy In Dialog no
 - Make Call Without Reg yes
 - Ans Call Without Reg yes
 - DNS SRV Auto Prefix no
- 7. In the **Subscriber Information**, enter **No_Number** in the **Display Name** field.
- 8. Enter the User ID (account number) and Password (passcode).

Note: Line 1 and Line 2 must have a different account number and passcode.

9. Change the value of the **Use Auth ID** field to **no**.

SIPLIPA				
technology, inc.			Sipura Phone A	Adapter Configuration
Info System SIP Provisio	oning Regional Line 1	Line 2	User 1 User 2	<u>User Login</u> <u>basic</u> advanced
Line Enable:	yes 💌			
Streaming Audio Server (SAS)				
SAS Enable:	no 💌		SAS DLG Refresh Intvl:	30
SAS Inbound RTP Sink:				
NAT Settings				\frown
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:	0×68		Network Jitter Level:	high 🔹
RTP TOS/DiffServ Value:	0×b8			
SIP Settings	5969			
SIF PORT	5060		Auth Decype Debect:	
SIR Debug Options	Dope	-	Audi Kesylic-Reboot:	yes -
Pestrict Source IP:			KTP LOG INON:	U
Result Source IP:				
Call Feature Settings				
Blind Attn-Xfer Enable:	no 💌		MOH Server:	
Xfer When Hangup Conf:	yes 💌			
Proxy and Registration				\frown
Proxy:	sip.primetalker.com		Use Outbound Proxy:	no 💌
Outbound Proxy:			Use OB Proxy In Dialog:	no 💌
Register:	yes 🔹		Make Call Without Reg:	yes 💌
Register Expires:	3600		Ans Call Without Reg:	yes 💌
Use DNS SRV:	no 💌		DNS SRV Auto Prefix:	no 🗾
Proxy Fallback Intvl:	3600			
Subscriber Information				
Display Name: 🧲	No Number		User ID:	0000000000
Password:	****		Use Auth ID:	no 💌
Auth ID:				
Mini Certificate:				
SRTP Private Key:				

Line 1 Configuration Page – Top

- 10. In the Audio Configuration section, change the value of the following fields:
 - Use Pref Codec Only no
 - G729a Enable yes
 - G723 Enable yes
 - G726-16 Enable no
 - G726-24 Enable no
 - G726-32 Enable no
 - G726-40 Enable no
 - DTMF Process INFO--no

- DTMF Process AVT yes
- DTMF Tx Method AVT

Supplementary Service Subscr	intion		
Call Waiting Serve	ves •	Block CID Serv:	ves •
Block ANC Serv:	ves •	Dist Ring Serv:	ves •
Cfwd All Serv:	ves •	Cfwd Busy Serv:	ves •
Cfwd No Ans Serv:	ves •	Cfwd Sel Serv:	ves •
Cfwd Last Serv:	ves •	Block Last Serv:	ves •
Accept Last Serv:	ves •	DND Serv:	ves •
CID Serv:	ves •	CWCID Serv:	ves •
Call Return Serv:	ves •	Call Back Serv:	ves •
Three Way Call Serv:	ves •	Three Way Coof Serv:	ves •
Atto Transfer Serv:	ves •	Upatto Transfer Serv:	ves •
MWI Serv:	ves •	VMWI Serv:	ves T
Speed Dial Serv:	ves •	Secure Call Serv:	ves •
Referral Serv:	ves •	Feature Dial Serv:	ves •
			() <u>_</u>
Audio Configuration			
Preferred Codec:	<u>2723</u>	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:	no 💌	Echo Supp Enable:	yes 💌
G726-24 Enable:	no 💌	FAX CED Detect Enable:	yes 💌
G726-32 Enable:	no 💌	FAX CNG Detect Enable:	yes 💌
G726-40 Enable:	no 💌	FAX Passthru Codec:	G711u 💌
DTMF Process INFO:	no 💌	FAX Codec Symmetric:	yes 💌
DTMF Process AVT:	yes 🔹	FAX Passthru Method:	NSE
DTMF T× Method:	AVT	FAX Process NSE:	yes 💌
Hook Flash Tx Method:	None 🖃	Release Unused Codec:	yes 💌
DistRing			
Dial Plan:	(*xx [3469]11 0 00 [2-9]xxxxxx	1xxx[2-9]xxxxxx xxxxxxxxxxxx.)	
Enable IP Dialing:	no 💌		
FXS Port Polarity Configuration			
Idle Polarity:	Forward	Caller Conn Polarity:	Reverse 💌
Callee Conn Polarity:	Reverse 💌		
	Undo All Changes	Submit All Changes	
User Login basic advanced			

Line 1 Configuration Page - Bottom

11. In the **Dial Plan** section, change the value of the **Dial Plan** field to match the dialing pattern used in your country (where you are using the PrimeTalker service).

For US calls, insert "1" as the first digit. For non-US calls, insert "011" before the rest of the phone number. For information on customizing your dial plan, refer to the SPA-2000 manual.

12. To save these settings, click the Submit All Changes button.