



Configuration Guide
for the
Sipura SPA-2000



CONNECT THE SPA-2000

Note:
Firmware 2.0.13(g) is required for service.

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.cafefone.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**.

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Sipura Phone Adapter Configuration

Info System SIP Provisioning Regional **Line 1** Line 2 User 1 User 2 [User Login](#) [basic](#) | [advanced](#)

Line Enable:

Streaming Audio Server (SAS)
SAS Enable: SAS DLG Refresh Intvl: 30
SAS Inbound RTP Sink:

NAT Settings
NAT Mapping Enable: NAT Keep Alive Enable:
NAT Keep Alive Msg: \$NOTIFY NAT Keep Alive Dest: \$PROXY

Network Settings
SIP TOS/DiffServ Value: 0x68 Network Jitter Level:
RTP TOS/DiffServ Value: 0xb8

SIP Settings
SIP Port: 5060 SIP 100REL Enable:
EXT SIP Port: Auth Resync-Reboot:
SIP Debug Option: RTP Log Intvl: 0
Restrict Source IP:

Call Feature Settings
Blind Attn-Xfer Enable: MOH Server:
Xfer When Hangup Conf:

Proxy and Registration
Proxy: Use Outbound Proxy:
Outbound Proxy: Use OB Proxy In Dialog:
Register: Make Call Without Reg:
Register Expires: 3600 Ans Call Without Reg:
Use DNS SRV: DNS SRV Auto Prefix:
Proxy Fallback Intvl: 3600

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

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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 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 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
Audio Configuration			
Preferred Codec:	G723	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 Passthu Codec:	G711u
DTMF Process INFO:	no	FAX Codec Symmetric:	yes
DTMF Process AVT:	yes	FAX Passthu Method:	NSE
DTMF Tx Method:	AVT	FAX Process NSE:	yes
Hook Flash Tx Method:	None	Release Unused Codec:	yes
Dial Plan			
Dial Plan:	(*xx[[3469]11 0 00][2-9]xxxxxx1xxx[2-9]xxxxxx xxxxxxxxxxxxx.)		
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			

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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 CafeFone 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.