

Sipura SPA- 1001 VoIP (ATA)Adapter Walk Th

Sipura SPA-1001 VoIP (ATA) Adapter Walk Through

Connect the ATA

- Connect the Network Cable from the MODEM to the ATA
- Connect the Telephone Handset to the ATA
- Connect the Power to the ATA

Power up and Configure

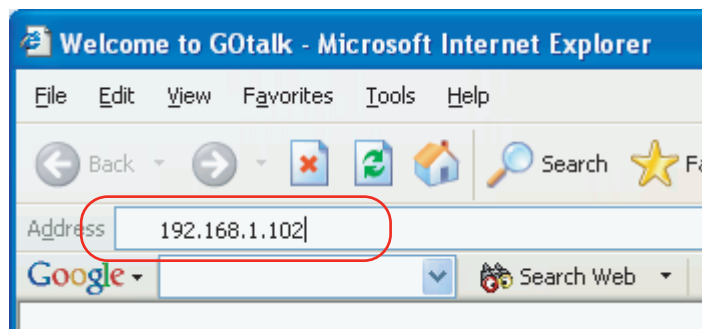
- Allow time for the ATA to Boot up ? About 60 Seconds
- Lift the receiver and press "****" (Star button 4 times)
- The IVR will kick in asking for an action.
- Type "110" then "#" (Hash-Pound key) on the number pad.
- Record the IP address that has been read back to you.

An example of this would be 192.168.1.102

Open Internet Explorer



Enter the IP address in to the Address line and select "Enter"



The default screen will appear

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Sipura Phone Adapter Configuration

Info | System | User 1 | User 2 | [Admin Login](#) | basic | [advanced](#)

System Information

DHCP:	Enabled	Current IP:	192.168.1.102
Host Name:	SipuraSPA	Domain:	
Current Netmask:	255.255.255.0	Current Gateway:	192.168.1.254
Primary DNS:	192.168.1.254		
Secondary DNS:			

Product Information

Product Name:	SPA-1001	Serial Number:	88012EA15916
Software Version:	2.0.12(SEa)	Hardware Version:	2.0.1(4864)
MAC Address:	000E08BA4AF0	Client Certificate:	Installed

System Status

Current Time:	1/1/2003 12:14:11	Elapsed Time:	00:09:37
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Select "Admin" login first

Sipura Phone Adapter Configuration

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

Current IP: 192.168.1.102

Then select "Advanced"

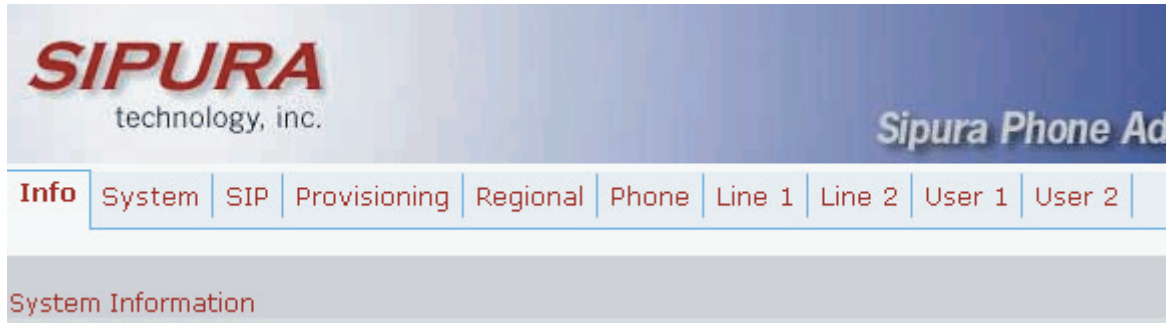
Sipura Phone Adapter Configuration

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

Current IP: 192.168.1.102



The following tabs will appear



Select the "System" Tab

Enter the following details here:

- Primary DNS 192.189.54.33
- Secondary DNS 192.189.54.17

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

System Configuration

Restricted Access Domains:

Enable Web Server: yes Web Server Port:

Enable Web Admin Access: yes Admin Passwd:

User Password:

Internet Connection Type

DHCP: yes

Static IP: NetMask:

Gateway:

Optional Network Configuration

HostName: Domain:

Primary DNS: Secondary DNS:

DNS Server Order: DNS Query Mode:

Syslog Server: Debug Server:

Debug Level: Primary NTP Server:

Secondary NTP Server:



Select Line 1 while in gAdvanced h Mode

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Sipura Phone Adapter Configuration

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

Enter the following Details further down in the page:

- **Sip server address** sip.gotalk.com
- **Register Expires** 240
- **Display name** phone number allocated to the customer
- **User ID** phone number allocated to the customer
- **Auth ID** uppercase 8 digit password allocated to the customer

SIP Settings

SIP Port:	5060	SIP 100REL Enable:	no
EXT SIP Port:	sip.gotalk.com	Auth Resync-Reboot:	yes
SIP Proxy-Require:	sip.gotalk.com	SIP Remote-Party-ID:	no
SIP Debug Option:	none	RTP Log Intvl:	0
Restrict Source IP:	no		

Call Feature Settings

Blind Attn-Xfer Enable:	no	MOH Server:	
Xfer When Hangup Conf:	yes	Conference Bridge URL:	
Conference Bridge Ports:	3		

Proxy and Registration

Proxy:	sip.gotalk.com	Use Outbound Proxy:	no
Outbound Proxy:	sip.gotalk.com	Use OB Proxy In Dialog:	yes
Register:	yes	Make Call Without Reg:	no
Register Expires:	240	Ans Call Without Reg:	no
Use DNS SRV:	no	DNS SRV Auto Prefix:	no
Proxy Fallback Intvl:	3600		

Subscriber Information

Display Name:	029037XXXX	User ID:	029037XXXX
Password:	*****	Use Auth ID:	no
Auth ID:			



Further Down the page Under "Audio Configuration"

Set the preferred codec to: G729a

Audio Configuration			
Preferred Codec:	<input type="text" value="G729a"/>	Silence Supp Enable:	<input type="text" value="no"/>
Use Pref Codec Only:	<input type="text" value="no"/>	Silence Threshold:	<input type="text" value="medium"/>
G729a Enable:	<input type="text" value="yes"/>	G723 Enable:	<input type="text" value="yes"/>
G726-16 Enable:	<input type="text" value="yes"/>	G726-24 Enable:	<input type="text" value="yes"/>
G726-32 Enable:	<input type="text" value="yes"/>	G726-40 Enable:	<input type="text" value="yes"/>
FAX Passthru Codec:	<input type="text" value="G711u"/>	DTMF Process INFO:	<input type="text" value="yes"/>
FAX Codec Symmetric:	<input type="text" value="yes"/>	DTMF Process AVT:	<input type="text" value="yes"/>
FAX Passthru Method:	<input type="text" value="NSE"/>	DTMF Tx Method:	<input type="text" value="Auto"/>
FAX Process NSE:	<input type="text" value="yes"/>	Hook Flash Tx Method:	<input type="text" value="None"/>
Release Unused Codec:	<input type="text" value="yes"/>		

At the bottom of the page select "Submit Change"

<input type="text" value="(*xx [3469]11 0 00 [2-9]xxxxxx 1xxx[2-9]xxxxxxS0 xxxxxxxxxxxxxx.)"/>	
<input type="text" value="no"/>	
<input type="button" value="Undo All Changes"/>	<input type="button" value="Submit All Changes"/>

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The following will appear:

SPA is updating your configuration. Unit may reset.
You will be redirected to the configuration page in 5 seconds.
If not, you can click [HERE](#) to return to the configuration page.

The device will then open up to the same page you were on.
The Configuration is now complete.

- Lift the telephone receiver and check for dial tone using the hand set.
- If you have dial tone you have completed setup and the service can be used.
- If you do not have dial tone. Carry out configuration again.

