



## **Application Note**

### **Asterisk BE with Remote Phones - Configuration Guide**

15 January 2009

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Tested versions: Ingate Firewall and SIParator version 4.6.4  
Startup Tool version 2.4.0  
Asterisk Business Edition version 2.1.1

## Revision History:

Revision	Date	Author	Comments
	2009-01-15	Scott Beer	Minor Edits

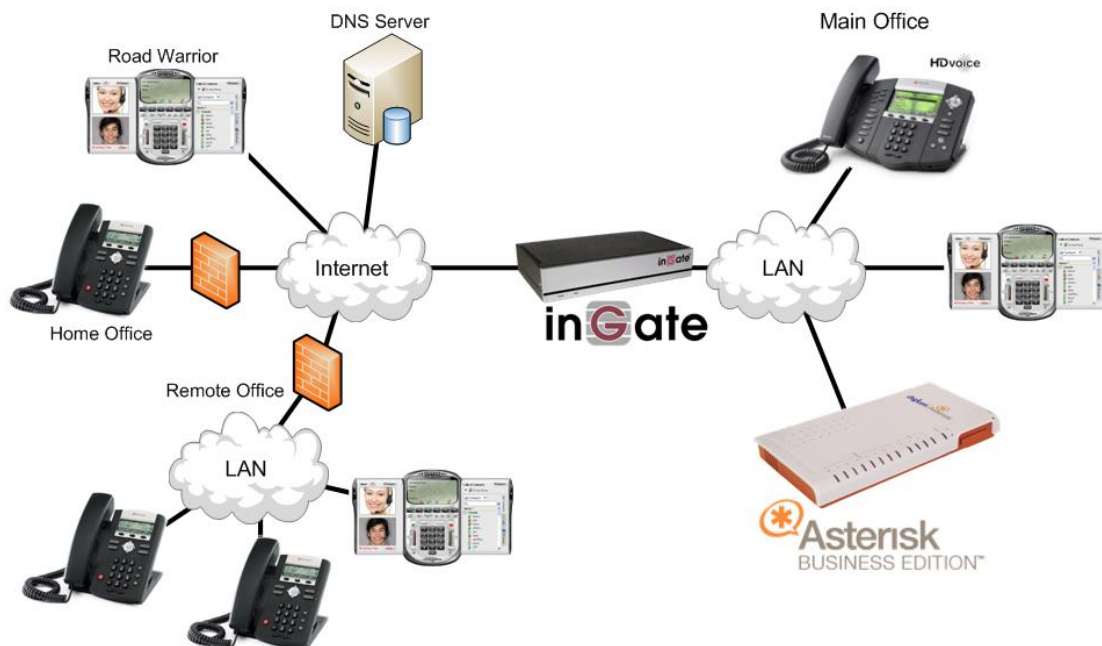
# 1 Asterisk Business Edition and Ingate

Digium offers Asterisk Business Edition, a professional-grade version of the Asterisk open source PBX, for the Linux operating system. Tailored for small and medium sized business applications, Asterisk Business Edition provides tested reliability of critical functions and features. It solves a wide range of challenges, from common PBX and key system replacements to highly-specialized applications. Asterisk Business Edition supports from 10 to 240 simultaneous calls per system.

The Asterisk Business Edition solution allows for the connectivity and use of a wide variety of SIP Phones, both desk phones and soft-phones. These SIP Phones can be from a number of different vendors, such as Polycom, Snom, Counterpath and more. These SIP Phones can be located both on the Enterprise LAN or abroad over the Internet, and in Remote/Home Offices.

Ingate offers SIParators and Firewalls, an Enterprise level SIP Session Border Controller (E-SBC) and SIP Security device. A powerful tool that offers enterprises a controlled and secured migration to VoIP (Voice over IP) and other live communications, based on Session Initiation Protocol (SIP). With the SIParator and Firewall, even the largest of businesses, with branch offices around the world and remote workers, can easily harness the productivity and cost-saving benefits of VoIP and other IP-based communications while maintaining current investments in security technology.

In this application, above and beyond the E-SBC capabilities that the Ingate products provide, the SIParator and Firewall are providing a number of additional features to enable remote SIP Phones connectivity to the Asterisk Business Edition IP-PBX solution. The Ingate products offer the use of the Remote SIP Connectivity Module, where there are features such as Far End NAT Traversal and a STUN Server. These features allow the Ingate to overcome NAT issues on the far end of the call.



## 1.1 Remote SIP Phone Support

In this application, the Asterisk Business Edition solution is the IP-PBX and SIP Domain Server. It is the call control server processing the phone features and PBX functionality required for an enterprise. It resides on the private LAN segment of enterprise, away from the Internet and protected by the Ingate from any malicious attacks.

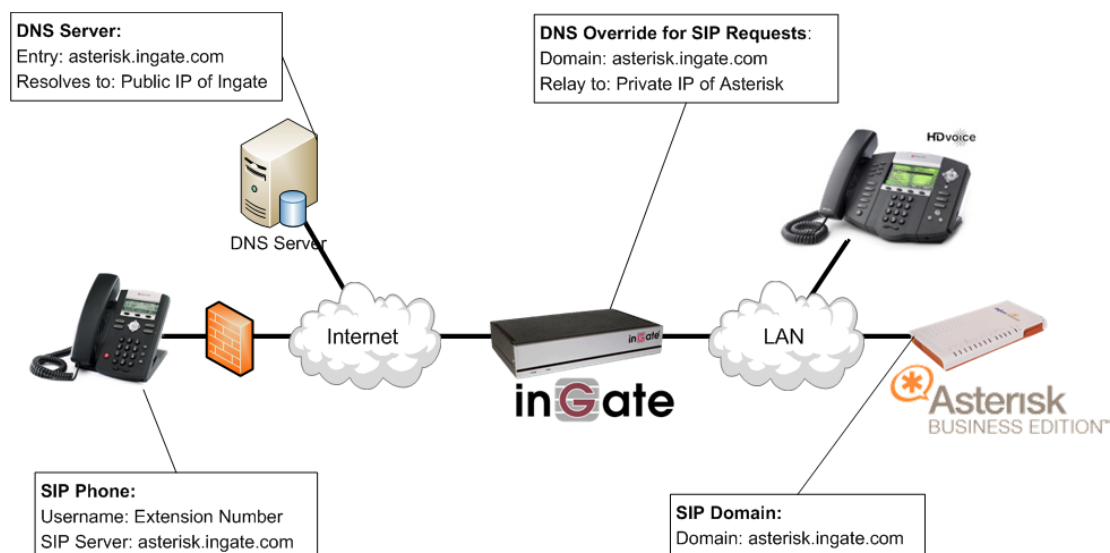
The Ingate SIParator or Firewall sits on the Enterprise network edge, providing a security solution for data and SIP communications with E-SBC functionality. It is responsible for all SIP communications security by providing Policy and Routing Rules to allow specific SIP traffic intended for the Enterprise.


The SIP Phones can be of any vendor type, located anywhere across the Internet or any remote networks.

### Requirements:

- 1) The use of a Fully Qualified Domain Name (FQDN) to resolve the SIP Domain of the Asterisk BE server. Meaning the Asterisk BE must respond to this SIP Domain, the SIP Phones must have this FQDN as the SIP Server address, all devices need to be able to do a DNS Lookup to resolve the FQDN to an IP address.
- 2) The Ingate must have the Remote SIP Connectivity Module to solve Far End NAT Traversal issues with remote phones.

### Application Diagram



Look for the Asterisk Business Edition Icon  to focus your attention to specific Asterisk BE setup instructions. These instructions are specific to the Ingate & Asterisk deployment with Remote SIP Phone.

## 2 Ingate Startup Tool

The Ingate Startup Tool is an installation tool for Ingate Firewall® and Ingate SIParator® products using the Ingate SIP Trunking module or the Remote SIP Connectivity module, which facilitates the setup of complete SIP trunking solutions or remote user solutions.

The Startup Tool is designed to simplify the initial “out of the box” commissioning and programming of the Network Topology, SIP Trunk deployments and Remote User deployments. The tool will automatically configure a user’s Ingate Firewall or SIParator to work with the Asterisk BE solution IP-PBX, this will setup all the routing needed to enable remote users to access and use the enterprise Asterisk BE IP-PBX. Thanks to detailed interoperability testing, Ingate has been able to create this tool with pre-configured setups for the Asterisk BE IP-PBX solutions with use with remote phones.

Download Free of Charge: The Startup Tool is free of charge for all Ingate Firewalls and SIParators. Get the latest version of the Startup Tool at [http://www.ingate.com/Startup\\_Tool.php](http://www.ingate.com/Startup_Tool.php)

For more detailed programming instructions consult the Startup Tool – Getting Started Guide, available here: [http://www.ingate.com/appnotes/Ingate\\_Startup\\_Tool\\_Getting\\_Started\\_Guide.pdf](http://www.ingate.com/appnotes/Ingate_Startup_Tool_Getting_Started_Guide.pdf)

Make sure that you always have the latest version of the configuration tool as Ingate continuously adds new vendors once interoperability testing is complete. If you don’t find your IP-PBX vendor or ITSP in the lists, please contact Ingate for further information.

The Startup Tool will install and run on any Windows 2000, Windows XP, Windows Vista, and Wine on Linux operating systems.

Keep in mind, this Ingate Startup Tool is a commissioning tool, not an alternate administration tool. This tool is meant to get an “out of the box” Ingate started with a pre-configured setup, enough to make your first call from Asterisk BE IP-PBX to any Remote SIP Phone. Additional programming and administration of this Ingate unit should be done through the Web Administration.

### 3 Connecting the Ingate Firewall/SIParator

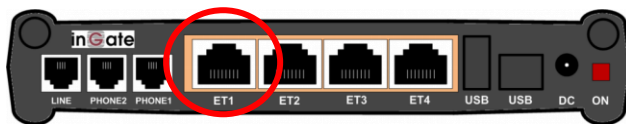
From the factory the Ingate Firewall and SIParator does not come preconfigured with an IP address or Password to administer the unit. Web administration is not possible unless an IP Address and Password are assigned to the unit via the Startup Tool or Console port.

The following will describe a process to connect the Ingate unit to the network then have the Ingate Startup Tool assign an IP Address and Password to the Unit.

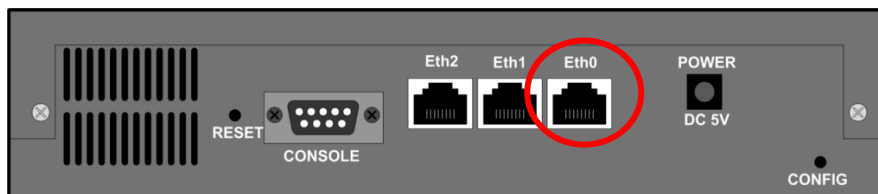
#### Configuration Steps:

- 1) Connect Power to the Unit.
- 2) Connect an Ethernet cable to “Eth0”. This Ethernet cable should connect to a LAN network. Below are some illustrations of where “Eth0” are located on each of the Ingate Model types. On SIParator SBE connect to “ET1”.

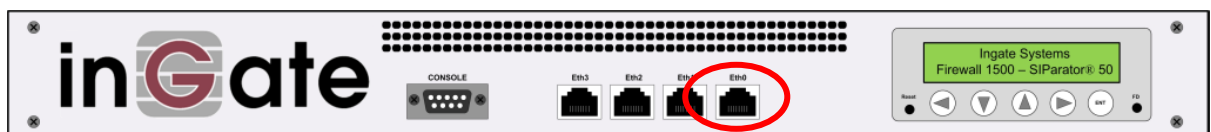
#### Ingate SIParator SBE (Back)



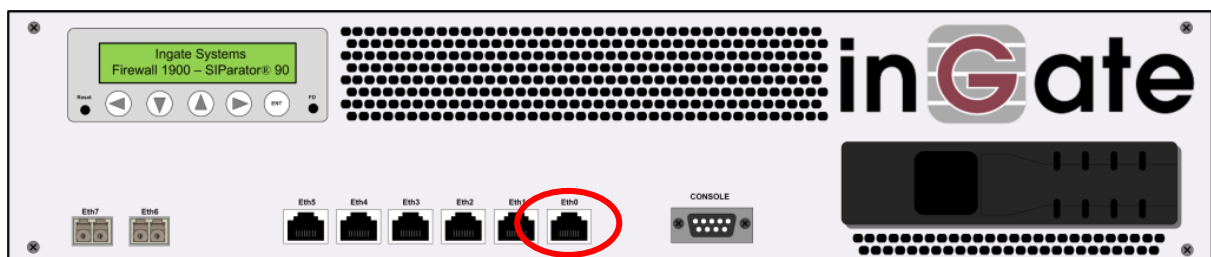
#### Ingate 1190 Firewall and SIParator 19 (Back)



#### Ingate 1500/1550/1650 Firewall and SIParator 50/55/65

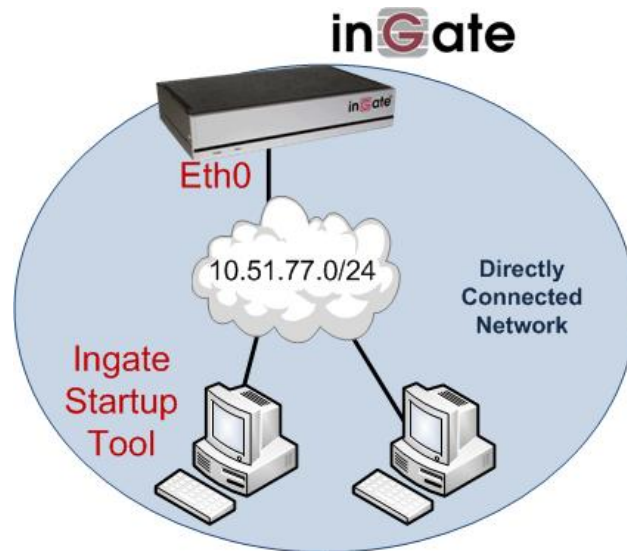


#### Ingate 1900 Firewall and SIParator 90



- 3) The PC/Server with the Startup Tool should be located on the same LAN segment/subnet. It is required that the Ingate unit and the Startup Tool are on the same LAN Subnet to which you are going to assign an IP Address to the Ingate Unit.

**Note:** When configuring the unit for the first time, avoid having the Startup Tool on a PC/Server on a different Subnet, or across a Router, or NAT device, Tagged VLAN, or VPN Tunnel. Keep the network Simple.



- 4) Proceed to Section 4: Using the Startup Tool for instructions on using the Startup Tool.

## 4 Using the Startup Tool

There are three main reasons for using the Ingate Startup Tool. First, the “Out of the Box” configuring the Ingate Unit for the first time. Second, is to change or update an existing configuration. Third, is to register the unit, install a License Key, and upgrade the unit to the latest software.

### 4.1 Configure the Unit for the First Time

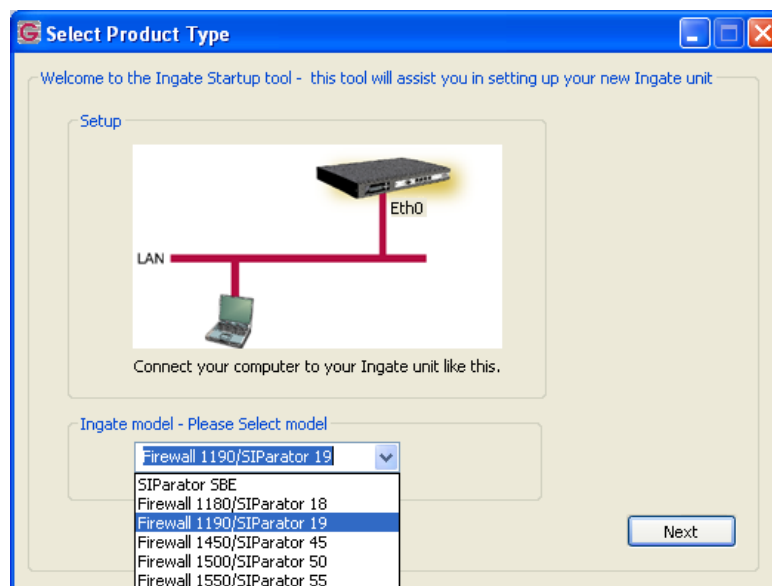
From the factory the Ingate Firewall and SIParator does not come preconfigured with an IP address or Password to administer the unit. Web administration is not possible unless an IP Address and Password are assigned to the unit via the Startup Tool or Console port.

In the Startup Tool, when selecting “Configure the unit for the first time”, the Startup Tool will find the Ingate Unit on the network and assign an IP Address and Password to the Ingate unit. This procedure only needs to be done ONCE. When completed, the Ingate unit will have an IP Address and Password assigned.

**Note:** If the Ingate Unit already has an IP Addressed and Password assigned to it (by the Startup Tool or Console) proceed directly to Section 4.2: “Change or Update Configuration”.

#### Configuration Steps:

- 1) Launch the Startup Tool
- 2) Select the Model type of the Ingate Unit, and then click Next.





- 3) In the “Select first what you would like to do”, select “Configure the unit for the first time”.

Ingate Startup Tool - Helps configure your Ingate unit

Ingate Startup Tool Version  
You are running the latest version of this tool.

Help

First select what you would like to do:

- ☒ Configure the unit for the first time
- ☐ Change or update configuration of the unit
- ☐ Check SIP configuration and logs
- ☐ Register this unit with Ingate
- ☐ Upgrade this unit
- ☒ Enable SIP module
- ☐ Configure Remote SIP Connectivity
- ☒ Configure SIP trunking
- ☐ Backup the created configuration
- ☐ Create a config without connecting to a unit
- ☐ This tool remembers passwords

Assign IP address and password, establish contact

Inside (Interface Eth0)

IP Address: 10 . 51 . 77 . 100

MAC Address: 00-d0-c9-a2-44-55

Select a password

Password: .....

Confirm Password: .....

Contact

Status

Ingate Startup Tool Version 2.4.0  
Startup tool version available on the Ingate web: 2.4.0  
You are running the latest version of the Startup tool.  
More information is available here: <http://www.ingate.com/startuptool.php>

- 4) Other Options in the “Select first what you would like to do”,

First select what you would like to do:

- ☒ Configure the unit for the first time
- ☐ Change or update configuration of the unit
- ☐ Check SIP configuration and logs
- ☐ Register this unit with Ingate
- ☐ Upgrade this unit
- ☒ Enable SIP module
- ☒ Configure Remote SIP Connectivity
- ☐ Configure SIP trunking
- ☐ Backup the created configuration
- ☐ Create a config without connecting to a unit
- ☐ This tool remembers passwords



- a. Select “Configure Remote SIP Connectivity” if you want the tool to configure Remote Phone access to the Asterisk Business Edition server.

- b. Select “Register this unit with Ingate” if you want the tool to connect with [www.ingate.com](http://www.ingate.com) to register the unit. If selected, consult the Startup Tool – Getting Started Guide.
  - c. Select “Upgrade this unit” if you want the tool to connect with [www.ingate.com](http://www.ingate.com) to download the latest software release and upgrade the unit. If selected, consult the Startup Tool – Getting Started Guide.
  - d. Select “Backup the created configuration” if you want the tool to apply the settings to an Ingate unit and save the config file.
  - e. Select “Creating a config without connecting to a unit” if you want the tool to just create a config file.
  - f. Select “The tool remembers passwords” if you want the tool to remember the passwords for the Ingate unit.
- 5) In the “Inside (Interface Eth0)”,
  - a. Enter the IP Address to be assigned to the Ingate Unit.
  - b. Enter the MAC Address of the Ingate Unit, this MAC Address will be used to find the unit on the network. The MAC Address can be found on a sticker attached to the unit.

Inside (Interface Eth0)

IP Address: 10 . 51 . 77 . 100

MAC Address: 00-D0-C9-A2-44-55

- 6) In the “Select a Password”, enter the Password to be assigned to the Ingate unit.

Select a password

Password: ••••••

Confirm Password: ••••••

- 7) Once all required values are entered, the “Contact” button will become active. Press the “Contact” button to have the Startup Tool find the Ingate unit on the network, assign the IP Address and Password.

Assign IP address and password, establish contact

Inside (Interface Eth0)

IP Address: 10 . 51 . 77 . 100

MAC Address: 00-D0-C9-A2-44-55

Select a password

Password: ••••••

Confirm Password: ••••••

Contact

- 8) Proceed to Section 4.3: Network Topology.

## 4.2 Change or Update Configuration

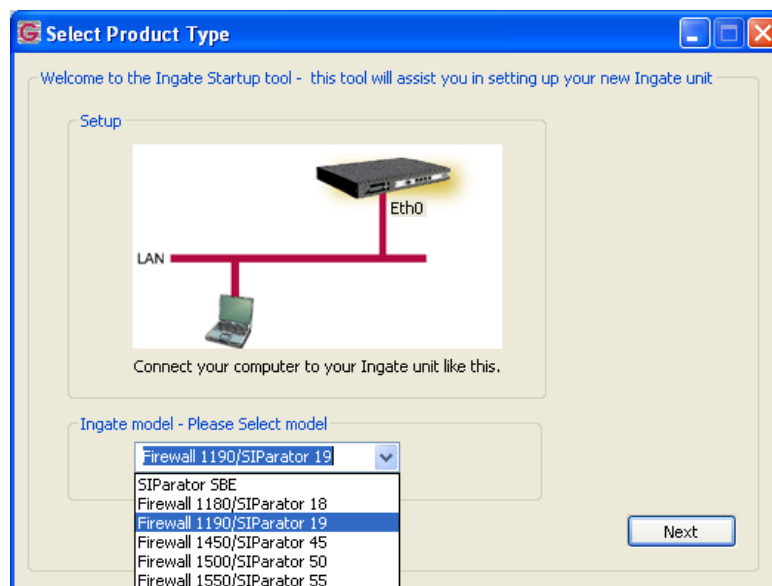
When selecting the “Change or update configuration of the unit” setting in the Startup Tool the Ingate Unit must have already been assigned an IP Address and Password, either by the Startup Tool – “Configure the unit for the first time” or via the Console port.

In the Startup Tool, when selecting “Change or update configuration of the unit”, the Startup Tool will connect directly with the Ingate Unit on the network with the provided IP Address and Password. When completed, the Startup Tool will completely overwrite the existing configuration in the Ingate unit with the new settings.

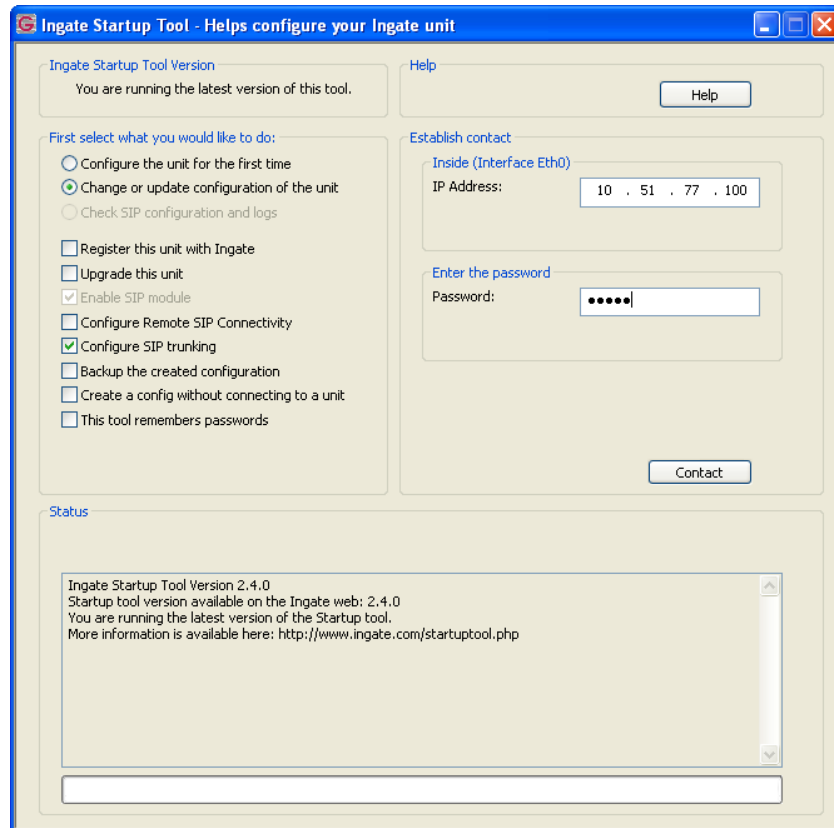
**Note:** If the Ingate Unit does not have an IP Addressed and Password assigned to it, proceed directly to Section 4.1: “Configure the Unit for the First Time”.

### Configuration Steps:

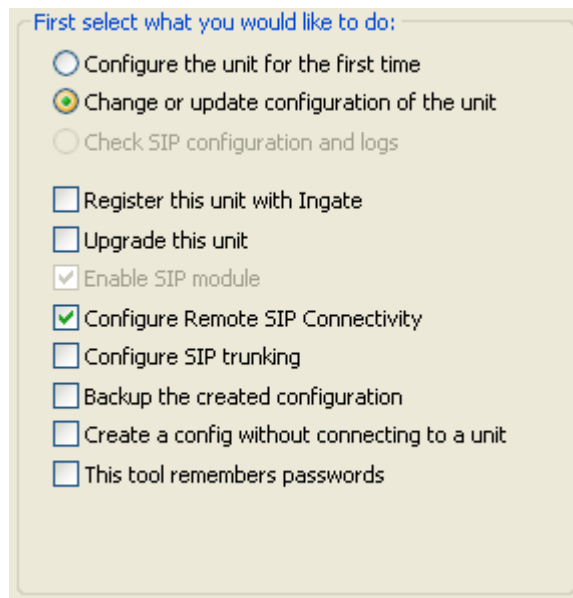
- 1) Launch the Startup Tool
- 2) Select the Model type of the Ingate Unit, and then click Next.



- 3) In the “Select first what you would like to do”, select “Change or update configuration of the unit”.

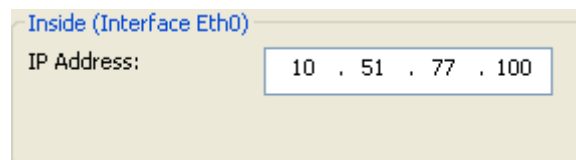


- 4) Other Options in the “Select first what you would like to do”,

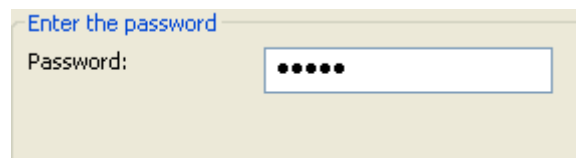


- a. Select “Configure Remote SIP Connectivity” if you want the tool to configure Remote Phone access to the Asterisk Business Edition server.

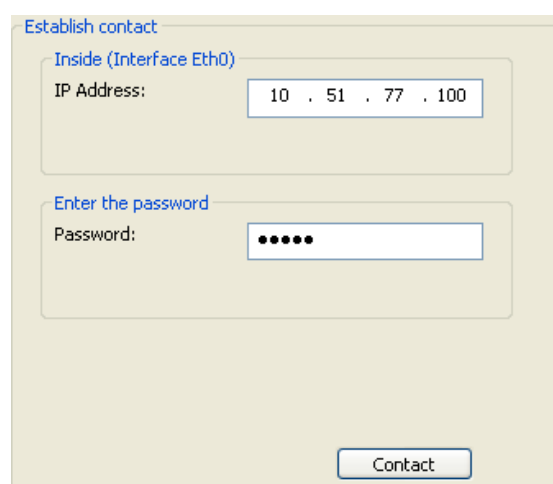
- b. Select “Register this unit with Ingate” if you want the tool to connect with [www.ingate.com](http://www.ingate.com) to register the unit. If selected, consult Startup Tool – Getting Started Guide.
  - c. Select “Upgrade this unit” if you want the tool to connect with [www.ingate.com](http://www.ingate.com) to download the latest software release and upgrade the unit. If selected, consult Startup Tool – Getting Started Guide.
  - d. Select “Backup the created configuration” if you want the tool to apply the settings to an Ingate unit and save the config file.
  - e. Select “Creating a config without connecting to a unit” if you want the tool to just create a config file.
  - f. Select “The tool remembers passwords” if you want the tool to remember the passwords for the Ingate unit.
- 5) In the “Inside (Interface Eth0)”,
- a. Enter the IP Address of the Ingate Unit.



- 6) In the “Enter a Password”, enter the Password of the Ingate unit.



- 7) Once all required values are entered, the “Contact” button will become active. Press the “Contact” button to have the Startup Tool contact the Ingate unit on the network.



- 8) Proceed to Section 4.3: Network Topology.

## 4.3 Network Topology

The Network Topology is where the IP Addresses, Netmask, Default Gateways, Public IP Address of NAT'ed Firewall, and DNS Servers are assigned to the Ingate unit. The configuration of the Network Topology is dependent on the deployment (Product) type. When selected, each type has a unique set of programming and deployment requirements, be sure to pick the Product Type that matches the network setup requirements.

The screenshot shows the 'Ingate Startup Tool' window with the 'Network Topology' tab selected. The 'Product Type' is set to 'Standalone SIParator'. The 'Inside (Interface Eth0)' section shows IP address '10 . 51 . 77 . 100' and Netmask '255 . 255 . 255 . 0'. The 'Outside (Interface Eth1)' section has 'Use DHCP to obtain IP' unchecked, IP address '172 . 51 . 77 . 100', Netmask '255 . 255 . 255 . 0', and Gateway '172 . 51 . 77 . 1'. A diagram shows the 'Ingate SIParator' connected to a 'LAN' with an 'IP-PBX' and an 'Existing firewall' connected to the 'Internet'. DNS server settings show Primary '4 . 2 . 2 . 2' and Secondary '0 . 0 . 0 . 0'. The status bar indicates 'Ingate Startup Tool Version 2.4.0, connected to: Ingate SIParator 19, IG-092-702-2122-0' and lists features like VoIP Survival, VPN, QoS, and licenses.

### Configuration Steps:

- 1) In the Product Type drop down list, select the deployment type of the Ingate Firewall or SIParator.

A close-up of the 'Product Type' dropdown menu, showing 'Standalone SIParator' selected.

**Hint:** Match the picture to the network deployment.

- 2) When selecting the Product Type, the rest of the page will change based on the type selected. Go to the Sections below to configure the options based on your choice. Select; Firewall, DMZ SIParator, DMZ-LAN SIParator, LAN SIParator, and Standalone SIParator.

### 4.3.1 Product Type: Firewall

When deploying an Ingate Firewall, there is only one way the Firewall can be installed. The Firewall must be the Default Gateway for the LAN; it is the primary edge device for all data and voice traffic out of the LAN to the Internet.

The screenshot shows the 'Ingate Startup Tool' window with the 'Network Topology' tab selected. The 'Product Type' is set to 'Firewall'. The 'Inside (Interface Eth0)' configuration shows an IP address of 10.51.77.1 and a netmask of 255.255.255.0. The 'Outside (Interface Eth1)' configuration has 'Use DHCP to obtain IP' checked, with an IP address of 12.23.34.45 and a netmask of 255.255.255.0. The gateway is set to 12.23.34.1. A diagram illustrates the firewall connecting the LAN (with an IP-PBX) to the Internet. DNS server settings are provided as 4.2.2.1 (Primary) and 4.2.2.2 (Secondary). The status section indicates the tool is connected to an Ingate Firewall 1190 and lists various features and the software version 4.6.2.

#### Configuration Steps:

- 1) In Product Type, select “Firewall”.

A close-up of the 'Product Type' dropdown menu. The text 'Product Type:' is followed by a dropdown box containing the word 'Firewall' and a downward-pointing arrow.

- 2) Define the Inside (Interface Eth0) IP Address and Netmask. This is the IP Address that will be used on the LAN side on the Ingate unit.

A close-up of the 'Inside (Interface Eth0)' configuration section. It shows two input fields: 'IP address:' with the value '10 . 51 . 77 . 1' and 'Netmask:' with the value '255 . 255 . 255 . 0'.

- 3) Define the Outside (Interface Eth1) IP Address and Netmask. This is the IP Address that will be used on the Internet (WAN) side on the Ingate unit.
  - a. A Static IP Address and Netmask can be entered
  - b. Or select “Use DHCP to obtain IP”, if you want the Ingate Unit to acquire an IP address dynamically using DHCP.

Outside (Interface Eth1)

☐ Use DHCP to obtain IP

IP Address: 12 . 23 . 34 . 45

Netmask: 255 . 255 . 255 . 248

☐ Allow https access to web interface from Internet

- 4) **Optional:** To configure Secure Web (https) from the Internet to the Ingate Unit for remote administration,
- Select “Allow https access to web interface from Internet”

Outside (Interface Eth1)

☐ Use DHCP to obtain IP

IP Address: 12 . 23 . 34 . 45

Netmask: 255 . 255 . 255 . 248

☒ Allow https access to web interface from Internet

- Create a Private Certificate for https access, enter the corresponding information required to generate a certificate.

Create certificate for https access

Common Name (CN): (Required) Your Name

Expire in (days): (Required) 365

Country Code (C): US

Organisation (O): Company Name

State/province(ST): NY

Organizational Unit(OU): Department

Email address: admin@email.com

Locality/town(L): Your City

OK Cancel

- 5) Enter the Default Gateway for the Ingate Firewall. The Default Gateway for the Ingate Firewall will always be an IP Address of the Gateway within the network of the outside interface (Eth1).

Gateway: 12 . 23 . 34 . 41

- 6) Enter the DNS Servers for the Ingate Firewall. These DNS Servers will be used to resolve FQDNs of SIP Requests and other features within the Ingate. They can be internal LAN addresses or outside WAN addresses.

DNS server

Primary: 4 . 2 . 2 . 1

Secondary: (Optional) 4 . 2 . 2 . 2



### 4.3.2 Product Type: Standalone

When deploying an Ingate SIParator in a Standalone configuration, the SIParator resides on a LAN network and on the WAN/Internet network. The Default Gateway for SIParator resides on the WAN/Internet network. The existing Firewall is in parallel and independent of the SIParator. Firewall is the primary edge device for all data traffic out of the LAN to the Internet. The SIParator is the primary edge device for all voice traffic out of the LAN to the Internet.

The screenshot shows the 'Ingate Startup Tool' window with the 'IP-PBX' tab selected. The 'Product Type' is set to 'Standalone SIParator'. The 'Inside (Interface Eth0)' section shows IP address '10 . 51 . 77 . 100' and Netmask '255 . 255 . 255 . 0'. The 'Outside (Interface Eth1)' section has 'Use DHCP to obtain IP' checked, with IP Address '12 . 23 . 34 . 45', Netmask '255 . 255 . 255 . 248', and Gateway '12 . 23 . 34 . 41'. A diagram shows the Ingate SIParator connected to a LAN with an IP-PBX and to the Internet via an Existing firewall. DNS server settings are Primary: '4 . 2 . 2 . 1' and Secondary: '4 . 2 . 2 . 2'. The Status section shows 'Ingate Startup Tool Version 2.4.0, connected to: Ingate SIParator 19, IG-092-702-2122-0' and a list of features: VoIP Survival, VPN, QoS, Enhanced Security, 10 SIP Traversal Licenses, 10 SIP User Registration Licenses, and Software Version: 4.6.2.

#### Configuration Steps:

- 1) In Product Type, select “Standalone SIParator”.

The close-up shows the 'Product Type:' label followed by a dropdown menu currently displaying 'Standalone SIParator'.

- 2) Define the IP Address and Netmask of the inside LAN (Interface Eth0). This is the IP Address that will be used on the Ingate unit to connect to the LAN network.

The close-up shows the 'Inside (Interface Eth0)' section with the IP address field set to '10 . 51 . 77 . 100' and the Netmask field set to '255 . 255 . 255 . 0'.

- 3) Define the Outside (Interface Eth1) IP Address and Netmask. This is the IP Address that will be used on the Internet (WAN) side on the Ingate unit.
  - a. A Static IP Address and Netmask can be entered
  - b. Or select “Use DHCP to obtain IP”, if you want the Ingate Unit to acquire an IP address dynamically using DHCP.

Outside (Interface Eth1)

☐ Use DHCP to obtain IP

IP Address: 12 . 23 . 34 . 45

Netmask: 255 . 255 . 255 . 248

☐ Allow https access to web interface from Internet

- 4) **Optional:** To configure Secure Web (https) from the Internet to the Ingate Unit for remote administration,
  - c. Select “Allow https access to web interface from Internet”

Outside (Interface Eth1)

☐ Use DHCP to obtain IP

IP Address: 12 . 23 . 34 . 45

Netmask: 255 . 255 . 255 . 248

☒ Allow https access to web interface from Internet

- d. Create a Private Certificate for https access, enter the corresponding information required to generate a certificate.

Create certificate for https access

Common Name (CN): Your Name

Expire in (days): 365

Country Code (C): US

Organisation (O): Company Name

State/province(ST): NY

Organizational Unit(OU): Department

Email address: admin@email.com

Locality/town(L): Your City

OK Cancel

- 5) Enter the Default Gateway for the Ingate SIParator. The Default Gateway for the SIParator will be the existing Firewalls IP Address on the DMZ network.

Gateway: 12 . 23 . 34 . 41

- 6) Enter the DNS Servers for the Ingate Firewall. These DNS Servers will be used to resolve FQDNs of SIP Requests and other features within the Ingate. They can be internal LAN addresses or outside WAN addresses.

### 4.3.3 Product Type: DMZ SIParator

When deploying an Ingate SIParator in a DMZ configuration, the Ingate resides on a DMZ network connected to an existing Firewall. The Ingate needs to know what the Public IP Address of the Firewall. This existing Firewall must be the Default Gateway for the DMZ network; the existing Firewall is the primary edge device for all data and voice traffic out of the LAN and DMZ to the Internet. SIP Signaling and Media must be forwarded to the Ingate SIParator, both from the Internet to the SIParator and from the DMZ to the LAN.

The screenshot shows the 'Ingate Startup Tool' window with the 'Network Topology' tab selected. The 'Product Type' is set to 'DMZ SIParator'. The 'DMZ (Interface Eth0)' section shows an IP address of 10.51.77.100 and a Netmask of 255.255.255.0. The 'LAN IP address range' section shows a Low IP of 192.168.1.1 and a High IP of 192.168.1.255. The 'Gateway' is 10.51.77.1 and the 'Firewall extern IP' is 12.23.34.45. A network diagram on the right shows the Internet connected to a Firewall, which is connected to the Ingate SIParator and the LAN. The LAN is connected to an IP-PBX. The 'DNS server' section shows a Primary of 4.2.2.2 and a Secondary of 4.2.2.1. The 'Status' section shows the Ingate Startup Tool Version 2.4.0, connected to Ingate SIParator 19, IG-092-702-2122-0. The status list includes VoIP Survival, VPN, QoS, Enhanced Security, 10 SIP Traversal Licenses, 10 SIP User Registration Licenses, and Software Version: 4.6.2.

#### Configuration Steps:

- 1) In Product Type, select “DMZ SIParator”.

The close-up shows the 'Product Type' dropdown menu with 'DMZ SIParator' selected.

- 2) Define the IP Address and Netmask of the DMZ (Interface Eth0). This is the IP Address that will be used on the Ingate unit to connect to the DMZ network side on the existing Firewall.

The close-up shows the 'DMZ (Interface Eth0)' section with the IP address field set to 192.168.100.100 and the Netmask field set to 255.255.255.0.

- 3) Define the LAN IP Address Range, the lower and upper limit of the network addresses located on the LAN. This is the scope of IP Addresses contained on the LAN side of the existing Firewall.

LAN IP address range

Low IP:	10 . 10 . 10 . 1
High IP:	10 . 10 . 10 . 255

- 4) Enter the Default Gateway for the Ingate SIParator. The Default Gateway for the SIParator will be the existing Firewall's IP Address on the DMZ network.

Gateway: 192 . 186 . 100 . 1

- 5) Enter the existing Firewall's external WAN/Internet IP Address. This is used to ensure correct SIP Signaling and Media traversal functionality. This is required when the existing Firewall is providing NAT.

Firewall extern IP: 98 . 87 . 76 . 65

- 6) Enter the DNS Servers for the Ingate Firewall. These DNS Servers will be used to resolve FQDNs of SIP Requests and other features within the Ingate. They can be internal LAN addresses or outside WAN addresses.

DNS server

Primary:	4 . 2 . 2 . 1
Secondary: (Optional)	4 . 2 . 2 . 2

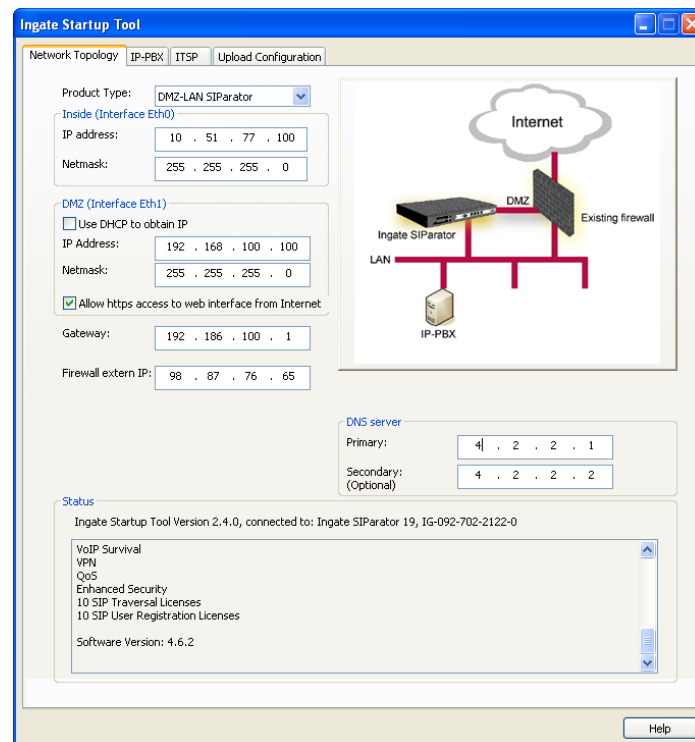
- 7) On the Existing Firewall, the SIP Signaling Port and RTP Media Ports need to be forwarded to the Ingate SIParator. The Ingate SIParator is an ICSA Certified network edge security device, so there are no security concerns forwarding network traffic to the SIParator.

On the existing Firewall:

- Port Forward the WAN/Internet interface SIP Signaling port of 5060 with a UDP/TCP Forward to the Ingate SIParator
- Port Forward the a range of RTP Media ports of 58024 to 60999 with a UDP Forward to the Ingate SIParator
- If necessary; provide a Rule that allows the SIP Signaling on port 5060 using UDP/TCP transport on the DMZ network to the LAN network
- If necessary; provide a Rule that allows a range of RTP Media ports of 58024 to 60999 using UDP transport on the DMZ network to the LAN network.

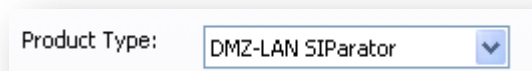
### 4.3.4 Product Type: DMZ-LAN SIParator

When deploying an Ingate SIParator in a DMZ-LAN configuration, the Ingate resides on a DMZ network connected to an existing Firewall and also on the LAN network. The Ingate needs to know what the Public IP Address of the Firewall. This existing Firewall must be the Default Gateway for the DMZ network; the existing Firewall is the primary edge device for all data and voice traffic out of the LAN and DMZ to the Internet. SIP Signaling and Media must be forwarded to the Ingate SIParator, from the Internet to the SIParator. The voice traffic from the LAN is directed to the SIParator then to the existing Firewall.

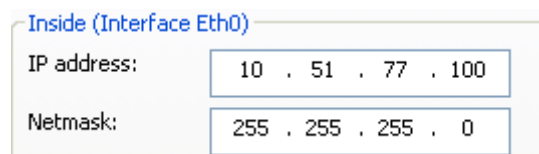


#### Configuration Steps:

- 1) In Product Type, select “DMZ-LAN SIParator”.

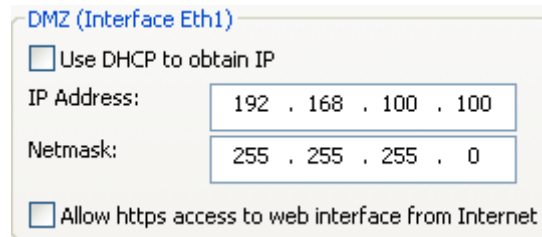


- 2) Define the IP Address and Netmask of the inside LAN (Interface Eth0). This is the IP Address that will be used on the Ingate unit to connect to the LAN network.



- 3) Define the IP Address and Netmask of the DMZ (Interface Eth1). This is the IP Address that will be used on the Ingate unit to connect to the DMZ network side on the existing Firewall.
  - a. A Static IP Address and Netmask can be entered

- b. Or select “Use DHCP to obtain IP”, if you want the Ingate Unit to acquire an IP address dynamically using DHCP.



DMZ (Interface Eth1)

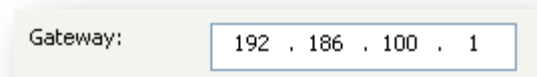
☐ Use DHCP to obtain IP

IP Address: 192 . 168 . 100 . 100

Netmask: 255 . 255 . 255 . 0

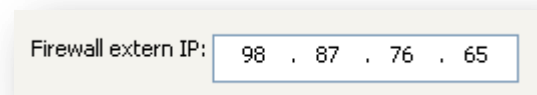
☐ Allow https access to web interface from Internet

- 4) Enter the Default Gateway for the Ingate SIParator. The Default Gateway for the SIParator will be the existing Firewall's IP Address on the DMZ network.



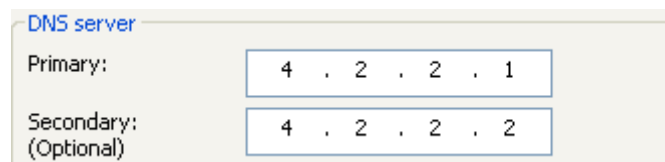
Gateway: 192 . 186 . 100 . 1

- 5) Enter the existing Firewall's external WAN/Internet IP Address. This is used to ensure correct SIP Signaling and Media traversal functionality. This is required when the existing Firewall is providing NAT.



Firewall extern IP: 98 . 87 . 76 . 65

- 6) Enter the DNS Servers for the Ingate Firewall. These DNS Servers will be used to resolve FQDNs of SIP Requests and other features within the Ingate. They can be internal LAN addresses or outside WAN addresses.



DNS server

Primary: 4 . 2 . 2 . 1

Secondary: (Optional) 4 . 2 . 2 . 2

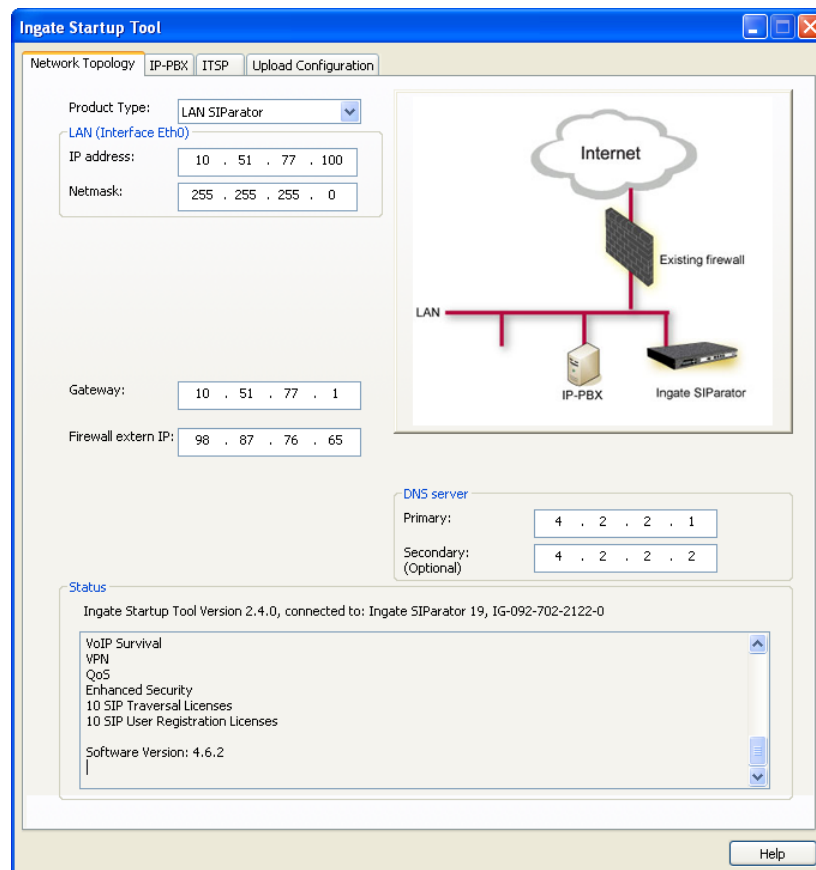
- 7) On the Existing Firewall, the SIP Signaling Port and RTP Media Ports need to be forwarded to the Ingate SIParator. The Ingate SIParator is an ICSA Certified network edge security device, so there are no security concerns forwarding network traffic to the SIParator.

On the existing Firewall:

- Port Forward the WAN/Internet interface SIP Signaling port of 5060 with a UDP/TCP Forward to the Ingate SIParator
- Port Forward the a range of RTP Media ports of 58024 to 60999 with a UDP Forward to the Ingate SIParator

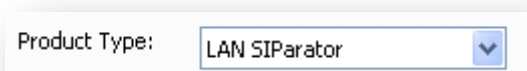
### 4.3.5 Product Type: LAN SIParator

When deploying an Ingate SIParator in a LAN configuration, the Ingate resides on a LAN network with all of the other network devices. The existing Firewall must be the Default Gateway for the LAN network; the existing Firewall is the primary edge device for all data and voice traffic out of the LAN to the WAN/Internet. SIP Signaling and Media must be forwarded to the Ingate SIParator, from the Internet to the SIParator. The voice traffic from the LAN is directed to the SIParator then to the existing Firewall.

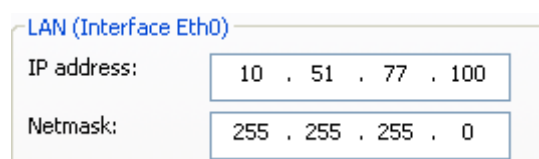


#### Configuration Steps:

- 1) In Product Type, select “LAN SIParator”.



- 2) Define the IP Address and Netmask of the inside LAN (Interface Eth0). This is the IP Address that will be used on the Ingate unit to connect to the LAN network.



- 3) Enter the Default Gateway for the Ingate SIParator. The Default Gateway for the SIParator will be the existing Firewall's IP Address on the DMZ network.

Gateway:	10 . 51 . 77 . 1
----------	------------------

- 4) Enter the existing Firewall's external WAN/Internet IP Address. This is used to ensure correct SIP Signaling and Media traversal functionality. This is required when the existing Firewall is providing NAT.

Firewall extern IP:	98 . 87 . 76 . 65
---------------------	-------------------

- 5) Enter the DNS Servers for the Ingate Firewall. These DNS Servers will be used to resolve FQDNs of SIP Requests and other features within the Ingate. They can be internal LAN addresses or outside WAN addresses.

DNS server	
Primary:	4 . 2 . 2 . 1
Secondary: (Optional)	4 . 2 . 2 . 2

- 6) On the Existing Firewall, the SIP Signaling Port and RTP Media Ports need to be forwarded to the Ingate SIParator. The Ingate SIParator is an ICSA Certified network edge security device, so there are no security concerns forwarding network traffic to the SIParator.

On the existing Firewall:

- Port Forward the WAN/Internet interface SIP Signaling port of 5060 with a UDP/TCP Forward to the Ingate SIParator
- Port Forward the a range of RTP Media ports of 58024 to 60999 with a UDP Forward to the Ingate SIParator



### 4.3.6 Product Type: LAN SIParator – “SBE SIParator Only”

This section is specific to the Ingate SBE SIParator when deploying in a LAN SIParator configuration, the Ingate SBE resides on a LAN network with all of the other network devices. The existing Firewall must be the Default Gateway for the LAN network; the existing Firewall is the primary edge device for all data and voice traffic out of the LAN to the WAN/Internet. SIP Signaling and Media must be forwarded to the Ingate SIParator, from the Internet to the SIParator. The voice traffic from the LAN is directed to the SIParator then to the existing Firewall.

The screenshot shows the 'Ingate Startup Tool' window with the 'Network Topology' tab selected. The 'Product Type' is set to 'LAN SIParator'. The 'LAN (Interface ET1)' section shows the IP address '10 . 51 . 77 . 200' and Netmask '255 . 255 . 255 . 0'. The 'Gateway' is '10 . 51 . 77 . 1'. The 'Firewall extern IP' is '98 . 87 . 76 . 65' and the 'Port range' is '58024 - 60999'. There is a checkbox for 'Allow DHCP Server, (setup in web GUI)'. The 'DNS server' section has 'Primary' as '4 . 2 . 2 . 1' and 'Secondary (Optional)' as '4 . 2 . 2 . 2'. A 'Status' box at the bottom indicates the tool is connected to the Ingate SIParator SBE and shows version information. A network diagram on the right illustrates the topology: Internet cloud connected to an 'Existing firewall', which is connected to a 'LAN' bus. The bus connects to an 'IP-PBX' and the 'Ingate SIParator'.

#### Configuration Steps:

- 1) In Product Type, select “LAN SIParator”.

A close-up of the 'Product Type' dropdown menu, showing 'LAN SIParator' selected.

- 2) Define the IP Address and Netmask of the inside LAN (Interface Eth0). This is the IP Address that will be used on the Ingate unit to connect to the LAN network.

A close-up of the 'LAN (Interface Eth0)' configuration fields. The 'IP address' is '10 . 51 . 77 . 100' and the 'Netmask' is '255 . 255 . 255 . 0'.

- 3) Enter the Default Gateway for the Ingate SIParator. The Default Gateway for the SIParator will be the existing Firewalls IP Address on the DMZ network.

Gateway:	<input type="text" value="10 . 51 . 77 . 1"/>
----------	---

- 4) Enter the existing Firewall's external WAN/Internet IP Address. This is used to ensure correct SIP Signaling and Media traversal functionality. This is required when the existing Firewall is providing NAT.

Firewall extern IP:	<input type="text" value="98 . 87 . 76 . 65"/>
---------------------	--

- 5) Enter a Port Range of media ports you need to configure the firewall to forward to the LAN SIParator

Port range:	<input type="text" value="58024"/>	-	<input type="text" value="60999"/>
-------------	------------------------------------	---	------------------------------------

- 6) Enter the DNS Servers for the Ingate Firewall. These DNS Servers will be used to resolve FQDNs of SIP Requests and other features within the Ingate. They can be internal LAN addresses or outside WAN addresses.

DNS server	
Primary:	<input type="text" value="4 . 2 . 2 . 1"/>
Secondary: (Optional)	<input type="text" value="4 . 2 . 2 . 2"/>

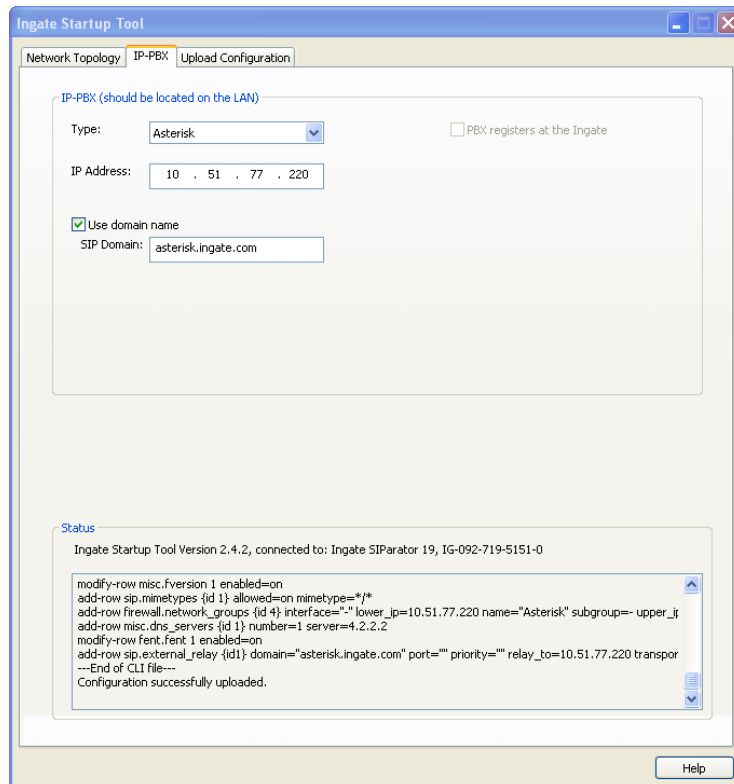
- 7) On the Existing Firewall, the SIP Signaling Port and RTP Media Ports need to be forwarded to the Ingate SIParator. The Ingate SIParator is an ICSA Certified network edge security device, so there are no security concerns forwarding network traffic to the SIParator.

On the existing Firewall:

- Port Forward the WAN/Internet interface SIP Signaling port of 5060 with a UDP/TCP Forward to the Ingate SIParator
- Port Forward the a range of RTP Media ports of 58024 to 60999 with a UDP Forward to the Ingate SIParator

## 4.4 IP-PBX

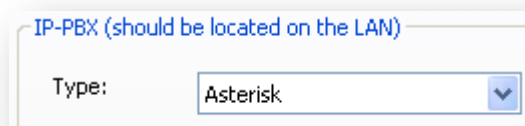
The IP-PBX section is where the IP Addresses and Domain location are provided to the Ingate unit. The configuration of the IP-PBX will allow for the Ingate unit to know the location of the Asterisk BE server as to direct SIP traffic for the use with the Remote Phones. The IP Address of the Asterisk BE server must be on the same network subnet at the IP Address of the inside interface of the Ingate unit. Ingate has confirmed interoperability with the Asterisk BE.



### Configuration Steps:



- 1) In the IP-PBX Type drop down list, select the appropriate IP-PBX vendor. Ingate has confirmed interoperability several of the leading IP-PBX vendors, the unique requirements of the vendor testing are contained in the Startup Tool. If the vendor choice is not seen, select “Generic PBX”.





- 2) Enter the IP Address of the Asterisk BE. The IP Address should be on the same LAN subnet as the Ingate unit.

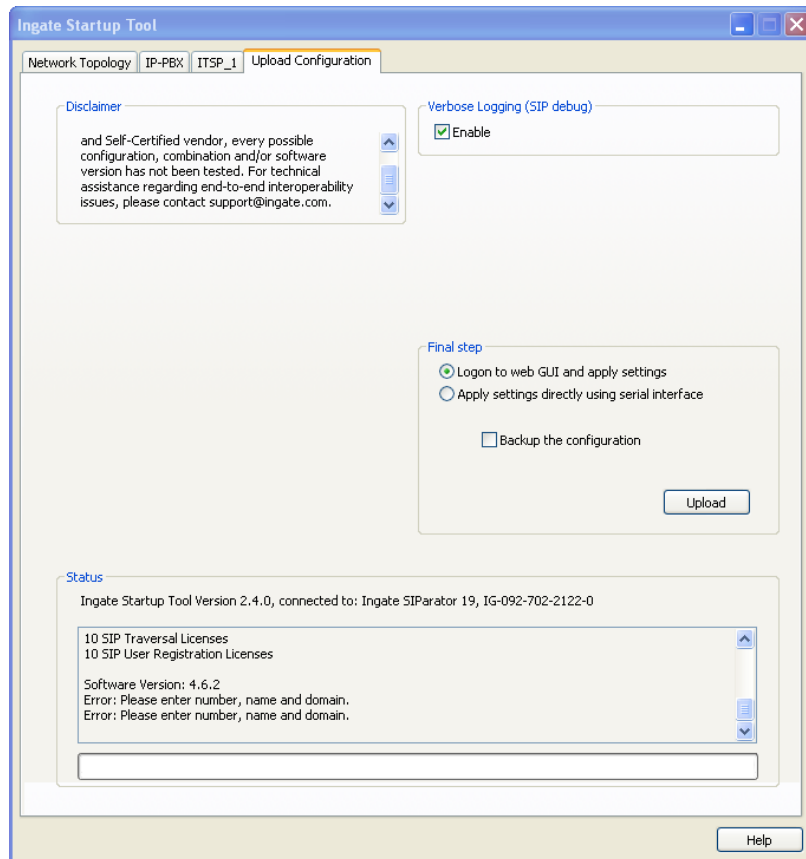
IP Address:

- 3) This solution requires the use of a FQDN for the SIP Domain of the Asterisk BE. This domain name is used to route SIP Requests to the Asterisk BE associated with that domain. Select “Use domain name” and enter the FQDN

☒ Use domain name  
SIP Domain:

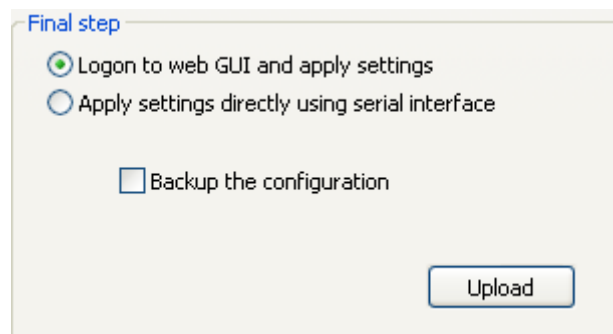
## 4.5 Upload Configuration

At this point the Startup Tool has all the information required to push a database into the Ingate unit. The Startup Tool can also create a backup file for later use.

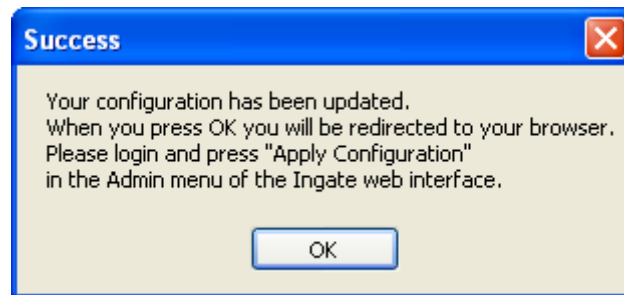


### Configuration Steps:

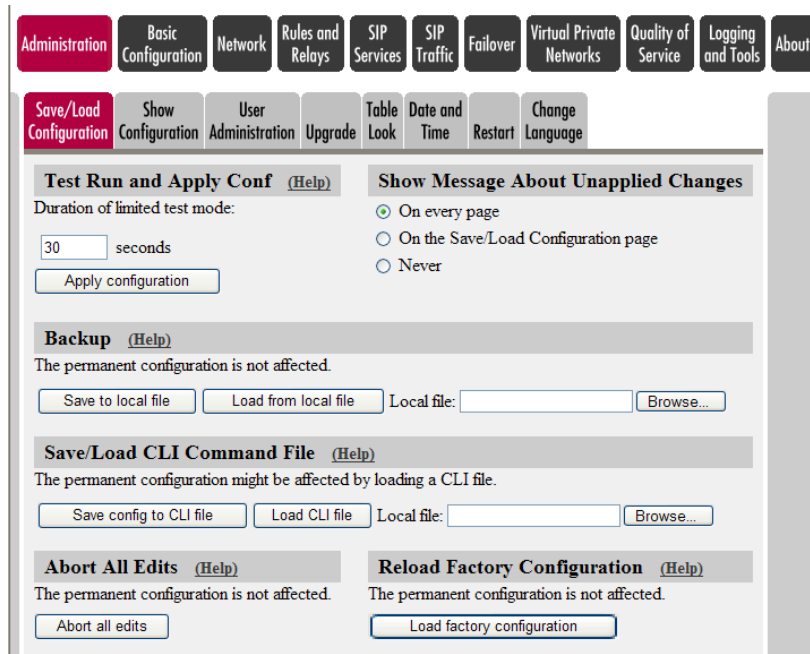
- 1) Press the “Upload” button. If you would like the Startup Tool to create a Backup file also select “Backup the configuration”. Upon pressing the “Upload” button the Startup Tool will push a database into the Ingate unit.



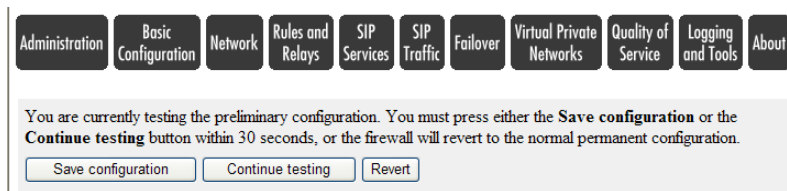
- 2) When the Startup has finished uploading the database a window will appear and once pressing OK the Startup Tool will launch a default browser and direct you to the Ingate Web GUI.



- 3) Although the Startup Tool has pushed a database into the Ingate unit, the changes have not been applied to the unit. Press "Apply Configuration" to apply the changes to the Ingate unit.



- 4) A new page will appear after the previous step requesting to save the configuration. Press "Save Configuration" to complete the saving process.



## 5 Asterisk Business Edition Setup



The Asterisk setup involves setting up the User Extensions, associating the SIP Domain, and adding an Outbound Proxy when using an Ingate SIParator.

### 5.1 User Extension on PBX Setup

Users is a shortcut for quickly adding and removing all the necessary configuration components for any new phone.

The screenshot shows the Asterisk Users interface. On the left is a sidebar with navigation links: System Status, Configure Hardware, mISDN Config, Trunks, Outgoing Calling Rules, Dial Plans, Users (selected), and Ring Groups. The main area is titled "User Extensions on PBX" and contains buttons for "Create New User", "Modify Selected Users", and "Delete Selected Users". Below these is a table titled "List of User Extensions" with columns: Extension, Full Name, Port, SIP, IAX, DialPlan, and OutBound CID. The table lists four users: 6000 (Scott Beer), 6001 (Test LAB1), 6002 (Remote Phone), and 6003 (Janne (Remote)). Each row has "Edit" and "Delete" buttons. A "Where to Buy" button is also present.

Extension	Full Name	Port	SIP	IAX	DialPlan	OutBound CID	
6000	Scott Beer	--	Yes	--	DialPlan1	none	Edit Delete
6001	Test LAB1	--	Yes	--	DialPlan1	none	Edit Delete
6002	Remote Phone	--	Yes	--	DialPlan1	none	Edit Delete
6003	Janne (Remote)	--	Yes	--	DialPlan1	none	Edit Delete

#### Configuration Steps:

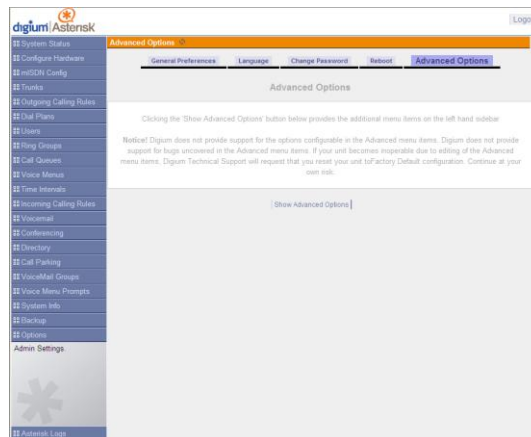
- 1) Select "Create New User" and the Asterisk BE will select the next available extension number and launch another screen.
- 2) In VoIP Settings section,
  - a. De-select NAT
  - b. Enter a Password for Authentication on the SIP Phone
- 3) The rest of the settings are phone or PBX specific features.

The screenshot shows the "Edit User Extension - 6002" configuration window. It has several sections: "General" with fields for Extension (6002), Name (Remote Phone), DialPlan (DialPlan1), CallerID (6002), and OutBound CallerID; "Enable Voicemail for this User" with a checked checkbox, VoiceMail Access PIN code (1234), and Email Address; "Technology" with checked SIP, IAX, and Analog Station options, and Codec Preference (First: u-law, Second: a-law, Third: G.729, Fourth: None, Fifth: None); "VoIP Settings" with MAC Address, Line Number (1), SIP/IAX Password (123456), NAT (unchecked), Can Reinvite (unchecked), DTMF Mode (RFC2833), insecure (no), and Other Options (3-Way Calling, In Directory, Call Waiting, CTI, Is Agent, Pickup Group: 1). At the bottom are "Cancel" and "Update" buttons.

## 5.2 SIP Settings

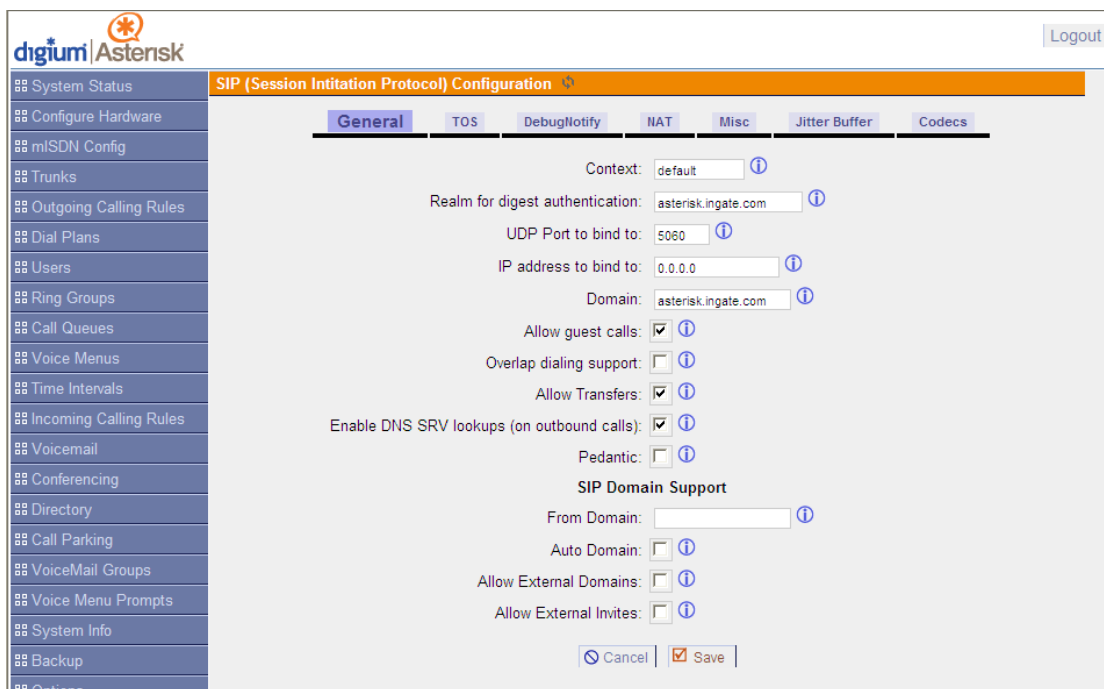
Clicking the 'Show Advanced Options' button below provides the advanced menu items on the left hand sidebar

**Note:** Digium does not provide support for the options configurable in the Advanced menu items. Digium does not provide support for bugs uncovered in the Advanced menu items. If your unit becomes inoperable due to editing of the Advanced menu items, Digium Technical Support will request that you reset your unit to Factory Default configuration. Continue at your own risk.



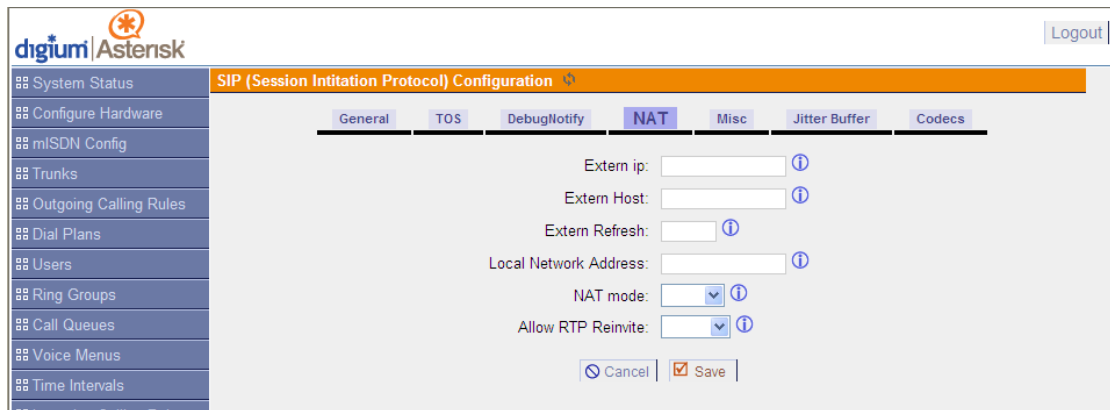
### Configuration Steps:

- 1) In the General configuration area
  - a. In the Domain field, enter the FQDN for the SIP Domain
  - b. In the Realm for digest authentication field, enter the FQDN of the SIP Domain





**Note:** The NAT area is left Blank



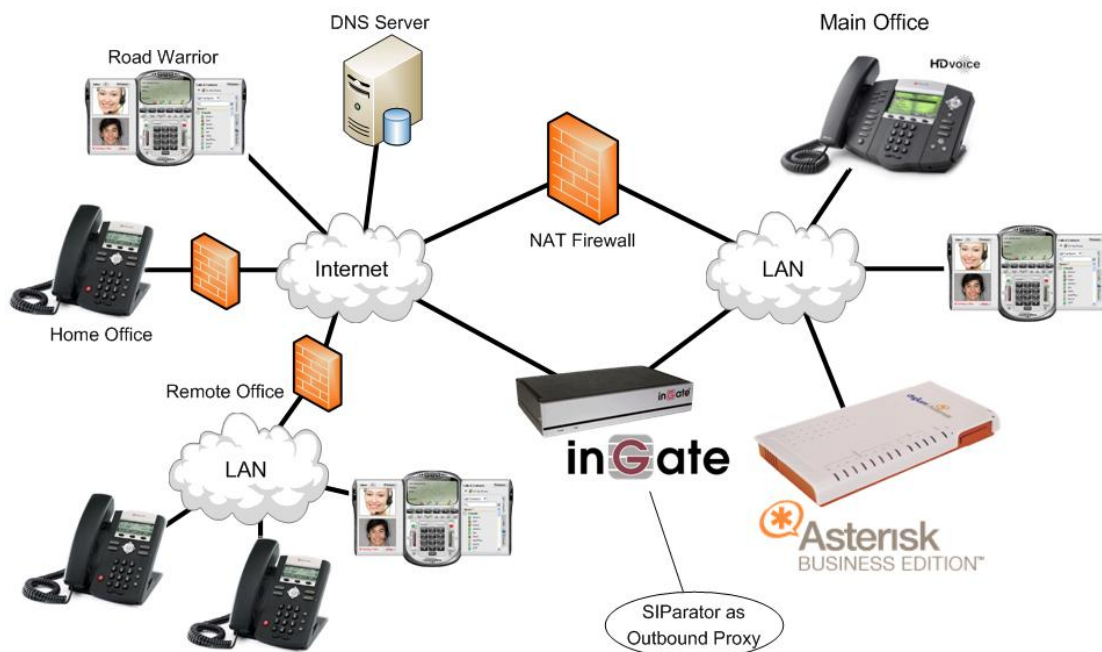
The screenshot shows the Asterisk SIP Configuration web interface. The left sidebar contains a menu with options: System Status, Configure Hardware, mISDN Config, Trunks, Outgoing Calling Rules, Dial Plans, Users, Ring Groups, Call Queues, Voice Menus, and Time Intervals. The main content area is titled 'SIP (Session Initiation Protocol) Configuration' and has several tabs: General, TOS, Debug/Notify, NAT (selected), Misc, Jitter Buffer, and Codecs. The NAT tab contains the following fields: 'Extern ip:' (text input), 'Extern Host:' (text input), 'Extern Refresh:' (text input), 'Local Network Address:' (text input), 'NAT mode:' (dropdown menu), and 'Allow RTP Reinvite:' (checkbox). At the bottom of the NAT tab are 'Cancel' and 'Save' buttons. The 'Save' button is checked.

### 5.3 Outbound Proxy Settings when using SIParator

Typically the Ingate SIParator is not Default Gateway of the network, so an Outbound Proxy must be configured in the Asterisk BE. Unfortunately, this feature isn't configurable through the GUI. However, you can set it up manually using the "outboundproxy" variable in /etc/asterisk/sip.conf. Use of this variable is fully documented inline within sip.conf.

The "outboundproxy" variable in /etc/asterisk/sip.conf is the Private IP Address of the Ingate SIParator.

This variable is not needed with the Ingate Firewall Product.



## 6 Troubleshooting

### 6.1 SIP Phone Registration Information

Now there are too many SIP Phone vendors to list here, so only an example of the CounterPath X-Lite will be shown.

The image displays three screenshots from the Asterisk management interface, with arrows indicating the flow of configuration and monitoring:

- Edit User Extension - 6000:** Shows the configuration for extension 6000. Key fields include: Extension: 6000, Name: Scott Beer, DialPlan: DiaPlan1, CallerID: 6000, OutBound CallerID: (empty), Enable Voicemail for this User: checked, VoiceMail Access PIN code: 1234, Email Address: scott@ingate.com, Technology: SIP, SIP/IAX Password: 123456, and 3-Way Calling: checked.
- Properties of Account 1:** Shows the account details for Scott Beer. Key fields include: Display Name: Scott Beer, User name: 6000, Password: (masked), Authorization user name: 6000, Domain: asterisk.ingate.com, and Dialing plan: #1|a|a.T;match=1;prestrip=2.
- SIP (Session Initiation Protocol) Configuration:** Shows the SIP configuration for the system. Key fields include: Context: default, Realm for digest authentication: asterisk.ingate.com, UDP Port to bind to: 5060, IP address to bind to: 0.0.0.0, Domain: asterisk.ingate.com, and SIP Domain Support: From Domain: (empty), Auto Domain: checked, Allow External Domains: checked, Allow External Invites: checked.
- SIP Traffic:** Shows the SIP Traffic status page. It includes a table for DNS Override for SIP Requests (Relay To) and a table for Registered Users (11 users).

**Note:** The successful registration of the SIP Phones can be seen in SIP Traffic -> SIP Status.

The screenshot shows the SIP Status page with the following information:

- Active Sessions (0 sessions):** There are no active sessions.
- Monitored SIP Servers:**

Monitored SIP Server	Port	Transport	Monitored SIP Server Status
ingate.com	5060	UDP	Monitored SIP server is online
10.51.77.60	5060	UDP	Monitored SIP server is online

- Registered Users (11 users):**

User	Registered From	Survival Aliases
11200@10.51.77.1	10.51.77.58	-
11201@10.51.77.1	10.51.77.59	-
11500@10.51.77.1	10.51.77.58	-

## 6.2 Startup Tool

### 6.2.1 Status Bar

Located on every page of the Startup Tool is the Status Bar. This is a display and recording of all of the activity of the Startup Tool, displaying Ingate unit information, software versions, Startup Tool events, errors and connection information. Please refer to the Status Bar to acquire the current status and activity of the Startup Tool.



### 6.2.2 Startup Tool - Configure Unit for the First Time

Right “Out of the Box”, sometimes connecting and assigning an IP Address and Password to the Ingate Unit can be a challenge. Typically, the Startup Tool cannot program the Ingate Unit. The Status Bar will display **“The program failed to assign an IP address to eth0”**.



### Possible Problems and Resolutions

Possible Problems	Possible Resolution
Ingate Unit is not Turned On.	Turn On or Connect Power (Trust me, I've been there)
Ethernet cable is not connected to Eth0.	Eth0 must always be used with the Startup Tool.
Incorrect MAC Address	Check the MAC address on the Unit itself. MAC Address of Eth0.
An IP Address and/or Password have already been assigned to the Ingate Unit	It is possible that an IP Address or Password have been already been assigned to the unit via the Startup Tool or Console

Possible Problems	Possible Resolution
Ingate Unit on a different Subnet or Network	The Startup Tool uses an application called “Magic PING” to assign the IP Address to the Unit. It is heavily reliant on ARP, if the PC with the Startup Tool is located across Routers, Gateways and VPN Tunnels, it is possible that MAC addresses cannot be found. It is the intension of the Startup Tool when configuring the unit for the first time to keep the network simple. See Section 3.
Despite your best efforts...	<ol style="list-style-type: none"> <li>1) Use the Console Port, please refer to the Reference Guide, section “Installation with a serial cable”, and step through the “Basic Configuration”. Then you can use the Startup Tool, this time select “Change or Update the Configuration”</li> <li>2) Factory Default the Database, then try again.</li> </ol>

### 6.2.3 Change or Update Configuration

If the Ingate already has an IP Address and Password assigned to it, then you should be able use a Web Browser to reach the Ingate Web GUI. If you are able to use your Web Browser to access the Ingate Unit, then the Startup should be able to contact the Ingate unit as well. The Startup Tool will respond with **“Failed to contact the unit, check settings and cabling”** when it is unable to access the Ingate unit.



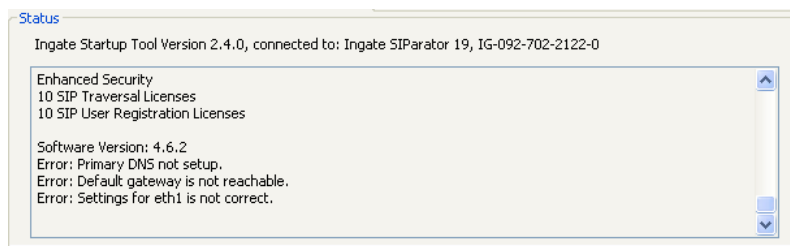
#### Possible Problems and Resolutions

Possible Problems	Possible Resolution
Ingate Unit is not Turned On.	Turn On or Connect Power
Incorrect IP Address	Check the IP Address using a Web Browser.
Incorrect Password	Check the Password.

Possible Problems	Possible Resolution
Despite your best efforts...	<ol style="list-style-type: none"> <li>1) Since this process uses the Web (http) to access the Ingate Unit, it should seem that any web browser should also have access to the Ingate Unit. If the Web Browser works, then the Startup Tool should work.</li> <li>2) If the Browser also does not have access, it might be possible the PC's IP Address does not have connection privileges in "Access Control" within the Ingate. Try from a PC that have access to the Ingate Unit, or add the PC's IP Address into "Access Control".</li> </ol>

## 6.2.4 Network Topology

There are several possible error possibilities here, mainly with the definition of the network. Things like IP Addresses, Gateways, NetMasks and so on.

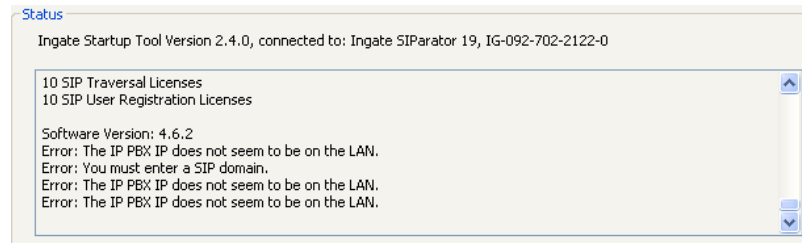


### Possible Problems and Resolutions

Possible Problems	Possible Resolution
Error: Default gateway is not reachable.	The Default Gateway is always the way to the Internet, in the Standalone or Firewall it will be the Public Default Gateway, on the others it will be a Gateway address on the local network.
Error: Settings for eth0/1 is not correct.	IP Address of Netmask is in an Invalid format.
Error: Please provide a correct netmask for eth0/1	Netmask is in an Invalid format.
Error: Primary DNS not setup.	Enter a DNS Server IP address

## 6.2.5 IP-PBX

The errors here are fairly simple to resolve. The IP address of the IP-PBX must be on the same LAN segment/subnet as the Eth0 IP Address/Mask.



### Possible Problems and Resolutions

Possible Problems	Possible Resolution
Error: The IP PBX IP does not seem to be on the LAN.	The IP Address of the IP-PBX must be on the same subnet as the inside interface of the Ingate Eth0.
Error: You must enter a SIP domain.	Enter a Domain, or de-select "Use Domain"
Error: As you intend to use RSC you must enter a SIP domain. Alternatively you may configure a static IP address on eth1 under Network Topology	Enter a Domain or IP Address used for Remote SIP Connectivity. Note: must be a Domain when used with SIP Trunking module.

## 6.3 Ingate Web GUI - Apply Configuration

At this point the Startup Tool has pushed a database to the Ingate Unit, you have Pressed "Apply Configuration" in Step 3) of Section 4.7 Upload Configuration, but the "Save Configuration" is never presented. Instead after a period of time the following webpage is presented. This page is an indication that there was a change in the database significant enough that the PC could no longer web to the Ingate unit.



## Possible Problems and Resolutions

Possible Problems	Possible Resolution
Eth0 Interface IP Address has changed	Increase the duration of the test mode, press “Apply Configuration” and start a new browser to the new IP address, then press “Save Configuration”
Access Control does not allow administration from the IP address of the PC.	Verify the IP address of the PC with the Startup Tool. Go to “Basic Configuration”, then “Access Control”. Under “Configuration Computers”, ensure the IP Address or Network address of the PC is allowed to HTTP to the Ingate unit.

## 6.4 DNS Benefits and Issues

As this solution is reliant on the resolution of a FQDN for the SIP Domain, the SIP Phones, the Ingate, and the Asterisk BE all need to be able to resolve the FQDN.

### DNS Standard Lookup

Ensure that SIP Phones, PCs and servers all have a DNS Server to which they can query a host name. There are some enterprises that have a internal DNS Server to manage internal host names.

PING tests using a domain is a good test to see if a network can resolve FQDNs.

### DYN DNS

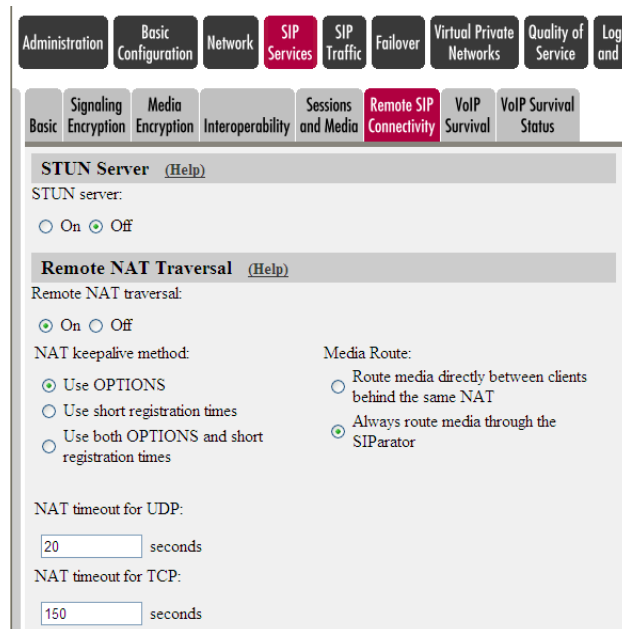
Dynamic DNS is a tool that can be use to provide smaller enterprises the ability to use a FQDN in a Dynamic Public IP environment. Visit [dyndns.org](http://dyndns.org) to get your free Domain name with Dynamic updating of the Enterprise IP address.

### DNS SRV Records

DNS Service Records offer the ability to do Load Balancing and Residency to any SIP Phone deployment. It offers the ability to use one FQDN and break the FQDN into multiple services, one for Web and another for SIP communications.

## 6.5 Remote SIP Connectivity Module

The Startup Tool will enable the Remote SIP Connectivity Module when selected. There are some additional options which help when running into some Far End NAT Traversal issues.



The screenshot shows the Asterisk SIP Configuration web interface. The top navigation bar includes tabs for Administration, Basic Configuration, Network, SIP Services (highlighted), SIP Traffic, Failover, Virtual Private Networks, Quality of Service, and Logging and Tracing. Below this, a sub-navigation bar includes Basic, Signaling Encryption, Media Encryption, Interoperability, Sessions and Media, Remote SIP Connectivity (highlighted), VoIP Survival, and VoIP Status. The main content area is divided into two sections: 'STUN Server' and 'Remote NAT Traversal'. The 'STUN Server' section has a 'STUN server:' label and two radio buttons: 'On' and 'Off', with 'Off' selected. The 'Remote NAT Traversal' section has a 'Remote NAT traversal:' label and two radio buttons: 'On' and 'Off', with 'On' selected. Below this, there are two columns of options. The left column is for 'NAT keepalive method:' with three radio buttons: 'Use OPTIONS' (selected), 'Use short registration times', and 'Use both OPTIONS and short registration times'. The right column is for 'Media Route:' with two radio buttons: 'Route media directly between clients behind the same NAT' and 'Always route media through the SIParator' (selected). At the bottom, there are two input fields: 'NAT timeout for UDP:' with a value of '20' seconds, and 'NAT timeout for TCP:' with a value of '150' seconds.

The SIParator can properly rewrite all the Remote SIP Phones SIP signaling. For this to work, the NAT box in front of the SIP client must keep the NAT hole open. In the SIParator there are two methods for doing this; using OPTIONS packets or using short registration times.

If the clients can respond to OPTIONS messages, you can use this method for keeping the NAT hole open. When this is used, the SIParator will send OPTIONS messages to the client with a frequency determined by the NAT timeout.

If the clients cannot respond to OPTIONS messages, you can instead use short registration times. This requires that the server accepts a registration time of the same length as the NAT timeout.

The NAT timeouts are used by the SIParator to determine how often OPTIONS or registrations should be sent.

In calls where the Remote NAT Traversal is used, the media (voice, video etc.) is usually routed through the SIParator. For two SIP clients behind the same remote NAT device, you can make the SIParator route media directly between the clients instead.