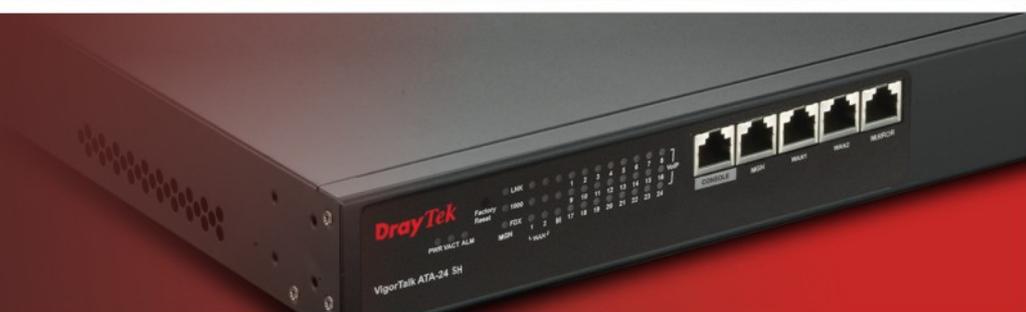


DrayTek

VigorTalk ATA-24 SH

24-Port Analogue Terminal Adapter

DrayTek



Your reliable networking solutions partner

User's Guide

V1.0

VigorTalk ATA-24 SH

24-Port Analogue Terminal Adapter

User's Guide

Version: 1.0

Date: 31/07/2009

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Table of Contents

1

Preface	1
1.1 LED Indicators and Connection	1
1.2 Hardware Installation	4
1.3 Power Connection.....	4
1.4 Management Port Connection	5
1.5 Default WAN 1 Port Connection.....	5
1.6 Detailed Explanation for the Connector	6

2

Configuring Basic Settings	9
2.1 Changing Password.....	9
2.2 Quick Setup.....	11
2.2.1 Adjusting WAN Connection Mode	11
2.2.2 Static Mode.....	12
2.2.3 DHCP Mode.....	14

3

Advanced Configuration.....	17
3.1 System setup	17
3.1.1 Status.....	17
3.1.2 Time.....	19
3.1.3 Syslog	21
3.1.4 Access Control.....	22
3.1.5 Configuration	23
3.1.6 Firmware Upgrade.....	24
3.1.7 Commit.....	26
3.1.8 Reboot	26
3.2 Network Setup.....	29
3.2.1 WAN and Internet Access Setup	29
3.2.2 MGN.....	35
3.3 Advanced Setup	36
3.3.1 Port Block	36
3.3.2 DDNS.....	37
3.4 Firewall Setup	39
3.4.1 DoS	39
3.5 VoIP Setup	42
3.5.1 Protocol.....	43
3.5.2 Port Settings	51
3.5.3 Speed Dial	54
3.5.4 Dial Plan	54

3.5.5 Tone Settings.....	56
3.5.6 Nat Traversal	59
3.5.7 Line Test	60
3.5.8 Miscellaneous	61
3.5.9 Incoming Call Barring	70
3.5.10 Statistics	73
3.5.11 Status.....	75
3.5.12 Call History	79
3.5.13 Configure Activate	80

4

Trouble Shooting81

4.1 Checking If the Hardware Status Is OK or Not.....	81
4.2 Checking If the Network Connection Settings on Your Computer Is OK or Not	82
4.3 Pinging the Adapter from Your Computer	85
4.4 Checking If the ISP Settings Are OK or Not.....	86
4.5 Backing to Factory Default Setting If Necessary	88
4.6 Contacting Your Dealer	88

Appendix A: Telnet Commands91

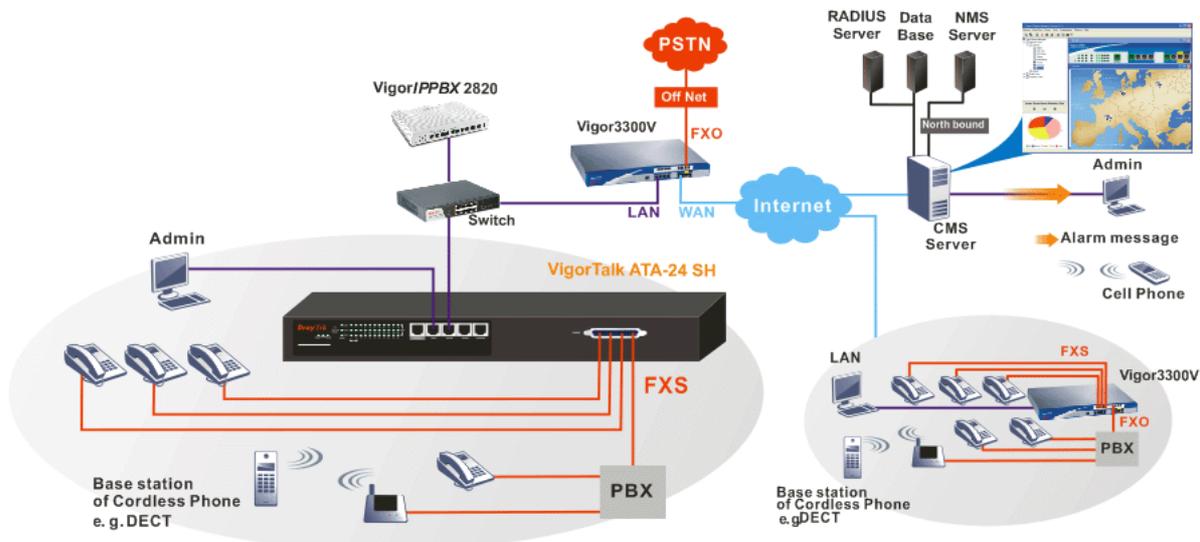
A.1 Introduction	91
A.2 Root Commands.....	91
A.2.1 Enter Function Commands	91
A.2.2 Other Commands	92
A.3 Advance Commands.....	92
A.3.1 General Commands	92
A.3.2 Port Block Commands.....	92
A.3.3 Portmirror Commands.....	92
A.4 Diagnostics Commands.....	93
A.4.1 General Commands	93
A.4.2 Learning_table Commands	93
A.4.3 Netstat Commands.....	93
A.4.4 Nslookup Commands	93
A.4.5 Ping Commands.....	93
A.4.6 Traceroute Commands.....	94
A.5 Firewall Commands.....	94
A.5.1 General Commands	94
A.5.2 DoS Commands	94
A.6 Network Commands	96
A.5.1 General Commands	96
A.5.2 MGN Commands.....	96
A.5.3 WAN Commands.....	97
A.6 System Commands	101
A.6.1 General Commands	101
A.6.2 View ARP Cache Table Command.....	101
A.6.3 View DHCP Assignment Command.....	101
A.6.4 View Routing Table Command	101
A.6.5 Administrator Control Commands.....	101

A.6.6 Auto Logout Commands.....	101
A.6.7 Config Commands.....	102
A.6.8 Manage Port Commands	102
A.6.9 Reboot Commands	103
A.6.10 Show Status Command.....	103
A.6.11 Syslogd Commands	103
A.6.13 Upgrade Commands	104
A.7 Voip Commands	105
A.7.1 General Commands	105
A.7.2 H248 Commands	105
A.7.3 Linetest Commands	106
A.7.4 MGCP Commands	106
A.7.5 Miscellaneous Commands	108
A.7.6 SIP Commands	111
A.7.7 Statistics Commands.....	116
A.7.8 VoIP Status Commands.....	117
A.7.9 Tone User Defined Commands.....	118
A.7.10 Config Commands.....	119
A.7.11 List Commands	120
A.7.12 Protocol Commands.....	120

1

Preface

The VigorTalk ATA-24 SH series integrates a rich suite of functions. These products are very suitable for providing multi-integrated solutions to SME markets. An application scenario for the VigorTalk ATA-24 SH is depicted in figure below, which illustrates interconnections among branch offices through the Internet via the VigorTalk ATA-24 SH adapter. Also, with Internet phone features, the company can benefit from reducing operation fees.



Internet Telephony, also known as Voice over Internet Protocol (VoIP), is a technology that allows you to make telephone calls using a broadband Internet connection instead of a regular (analog) phone line. It offers features and services that are unavailable with a traditional phone at no additional cost. Because Internet Telephony requires strictly minimal packet delay and jitter (since voice quality is intolerant of packet loss), the adapter integrates VoIP feature with QoS and packet loss concealment mechanisms to effectively transport high priority voice traffic over IP with low latency. Another feature is T.38 fax relay. By enabling and configuring fax rate on a dial peer, the originating and the terminating adapter can enter fax relay transfer mode. By using the T.38 function, customers can also save on fax expenses.

1.1 LED Indicators and Connection

The VigorTalk ATA-24 SH has 2 WAN interfaces. Each interface can be connected to an individual Internet Service Provider. The VigorTalk ATA-24 SH also supports a backup function for WAN interfaces – a user can select one WAN interface to be a backup interface. If the master interface fails, the backup interface will take the place of the master interface immediately.

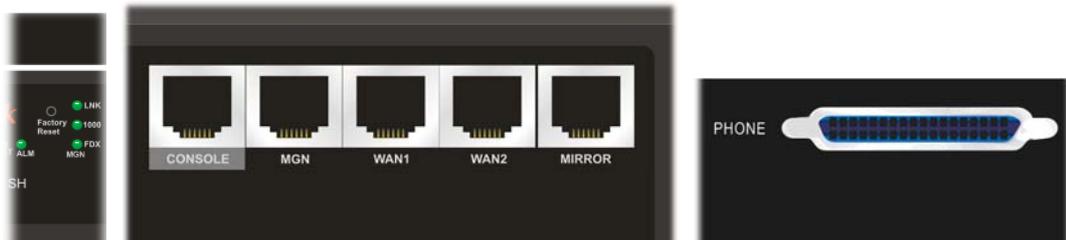


LED	Status	Explanation	
PWR	On	The adapter is powered on.	
	Off	The adapter is powered off.	
VACT	On/Blinking	The system is active.	
	Off	The system is hanged.	
ALM	On	The system alarm is active.	
MGN	LNK	On	The Ethernet link is established.
		Blinking	The data transmission is done through the corresponding port.
		Off	No Ethernet link is established.
	1000	On	It means that a normal 1000 Mbps connection is through its corresponding port.
		Off	It means that a normal 100/10 Mbps connection is through its corresponding port.
	FDX	On	It means a full duplex connection on corresponding port.
		Off	It means a half duplex connection on corresponding port.
	WAN/MIRROR	LNK	On
Off			No Ethernet link is established.
1000		On	It means that a normal 1000 Mbps connection is through its corresponding port.
		Off	It means that a normal 100 Mbps connection is through its corresponding port.
FDX		On	It means a full duplex connection on corresponding port.
		Off	It means a half duplex connection on corresponding port.

LED	Status	Explanation
VoIP (1-24)	On	The phone is off hook (the handset of phone is hanging).
	Blinking	A phone call is incoming or on-line.

Factory Reset:

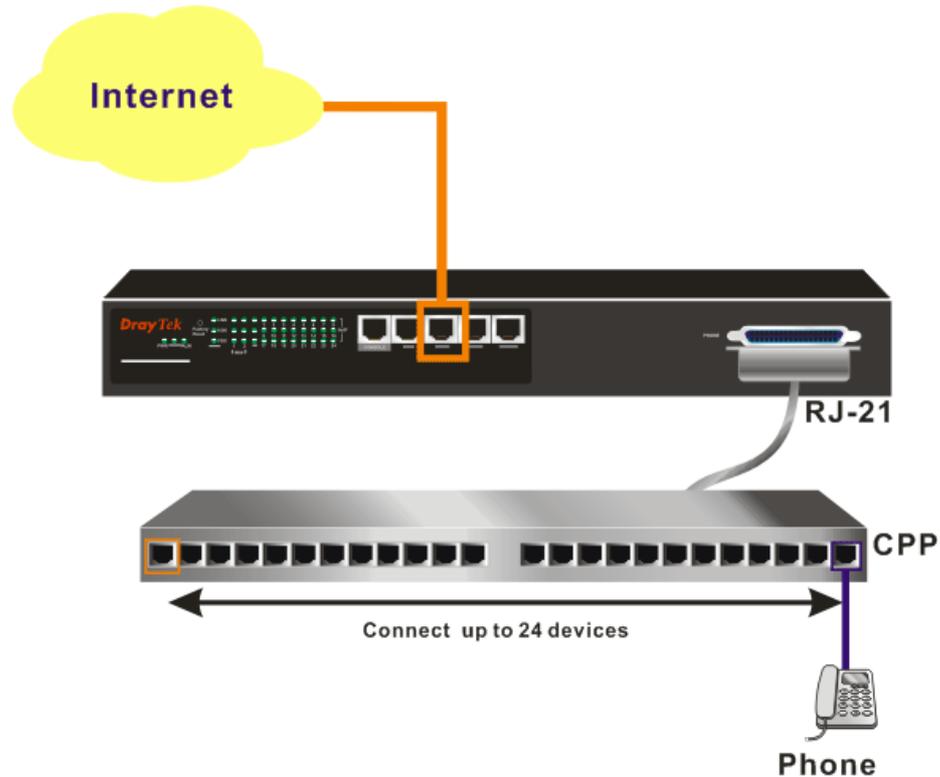
Used to restore the default settings. Turn on the adapter (**VACT LED** is blinking). Press the hole and hold for more than 5 seconds. When you see the **VACT LED** begins to blink rapidly than usual, release the button. Then the adapter will restart with the factory default configuration.



Interface	Description
Factory Reset	Used to restore the default settings. Turn on the adapter (VACT LED is blinking). Press the hole and hold for more than 5 seconds. When you see the VACT LED begins to blink rapidly than usual, release the button. Then the adapter will restart with the factory default configuration.
CONSOLE	Provided for technician use.
MGN	Connector for local management.
WAN (WAN1 ~ WAN2)	Connector for remote networked devices.
MIRROR	Connector for security monitor.
PHONE	Connected to telephones.

1.2 Hardware Installation

Phone service is available in VigorTalk ATA-24 SH. You can connect a RJ-11 jack port attached in a CPP to a telephone directly. The following example shows the connection for *Voice transmission* through CPP.



1.3 Power Connection

Before you purchasing the device, please check your environment to determine which power type that matches with your requirement.

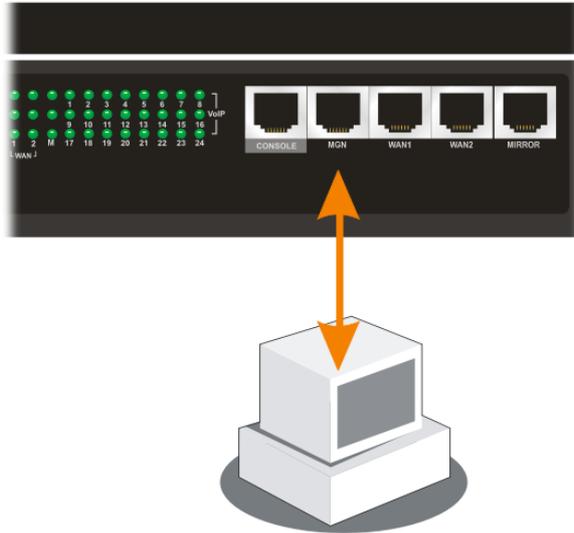
The AC input and ground connections can be done on the rear panel. You can connect the rack to ground by using spring screws.



1. Connect the female end of the power cord to the power socket on the rear panel of VigorTalk ATA-24 SH.
2. Connect the other end of the cord to a power outlet and make sure that no objects obstruct the airflow of the fans (located on the rear side of the unit).

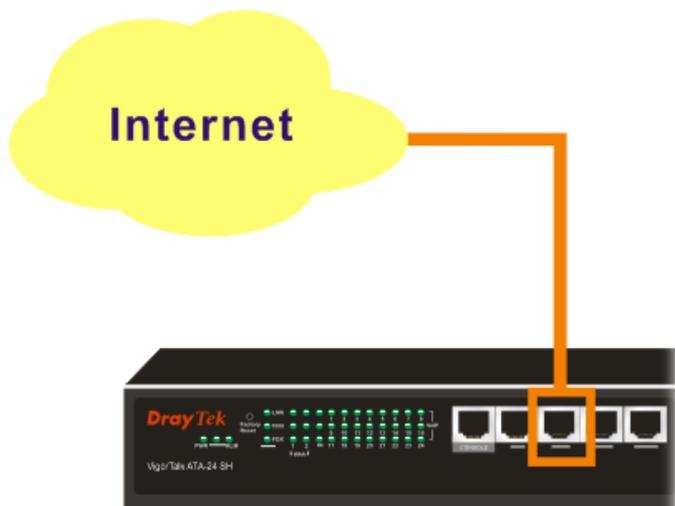
1.4 Management Port Connection

Users can connect the RJ-45 cable to the MGN port of the device. The IP address is **192.168.1.1** by default. The subnet of PC should be the same as default IP setting.



1.5 Default WAN 1 Port Connection

Users can connect the RJ-45 cable to the WAN1 port of the device. The IP address is **172.16.1.2** by default. Modify the IP address of the network segment you want to use for accessing into Internet if necessary.



1.6 Detailed Explanation for the Connector

Here provides you detailed explanation for some specific connectors that you have to be familiar.

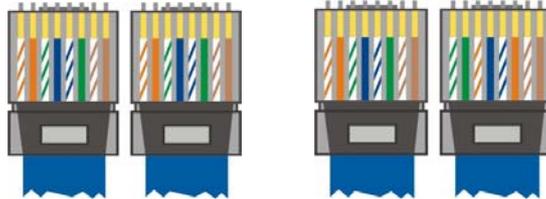
The RS232 Connector

The RJ45 connection jet is used for CLI commands for system configuration and control functions in the VigorTalk ATA-24. The jet is used for initialization of the VigorTalk ATA-24 SH during preliminary installation. The “management cable”, as shown below, converts the RJ45 to the RS232 interface. The RJ45 jet connects to a console interface in the VigorTalk ATA-24, while the RS232 DB9 connects to a console port on the computer. The default setting of the console port is “**baud rate 115200, no parity, and 8 bit with 1 stop bit.**”



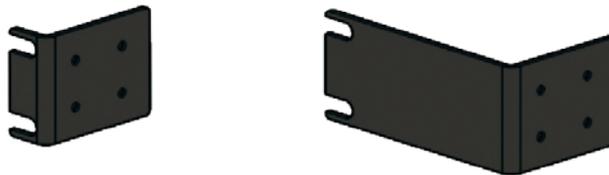
Standard 10/100 Base-T Ethernet Interface Connector

RJ45 jets provide 10/100 Base-T Ethernet interfaces. The interface supports MDI/MDIX auto-detection of either straight or crossover RJ45 cables. These cables are used on WAN, LAN, and DMZ interfaces.

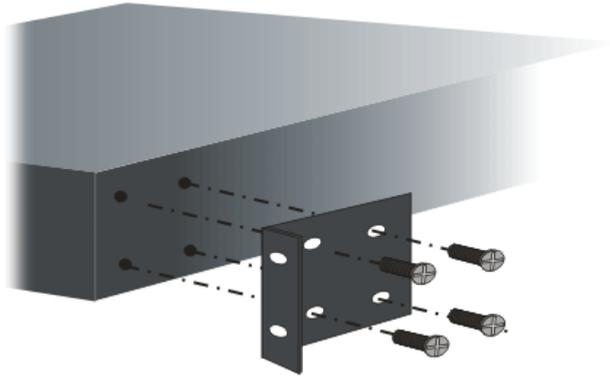


Chassis Connections

The VigorTalk ATA-24 SH can be mounted on a rack by using standard brackets in a 19-inch rack or optional larger brackets on 23-inch rack (not included). The bracket for 19- and 23-inch racks are shown below.



Attach the brackets to the chassis of a 19- or a 23-inch rack (as shown in the figures below). Repeat the above procedure for the second bracket, which attaches the other side of the chassis.



After the bracket installation, the VigorTalk ATA-24 SH chassis can be installed in a rack by using four screws for each side of the rack.



Desktop Type Installation

Rubber pads are included with the VigorTalk ATA-24SH. These rubber pads improve the air circulation and decrease unnecessary rubbing on the desktop.

2

Configuring Basic Settings

For use the adapter properly, it is necessary for you to change the password of web configuration for security and adjust primary basic settings.

This chapter explains how to setup a password for an administrator and how to adjust basic settings for accessing Internet successfully.

2.1 Changing Password

To change the password for this device, you have to access into the web browser with default password first.

1. Make sure your computer connects to the adapter MGN port correctly.

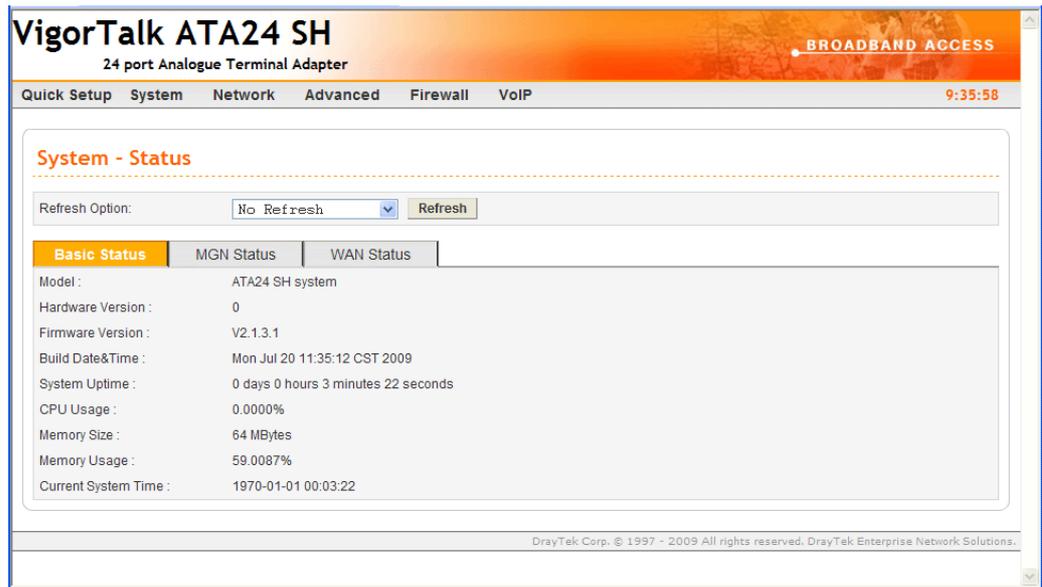


Notice: You may set up the IP address of the computer to be the same subnet as **the default IP address of Vigor adapter 192.168.1.1**. For the detailed information, please refer to the later section - Trouble Shooting of this guide.

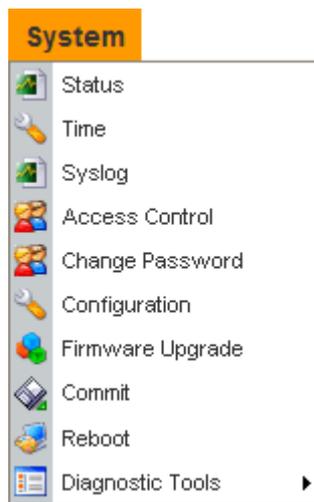
2. Open a web browser on your PC and type **http://192.168.1.1**. A pop-up window will open to ask for username and password. Please type default values on the window for the first time accessing. The default value for user name is “**admin**” and the password is “**1234**”. Next, click **OK**.



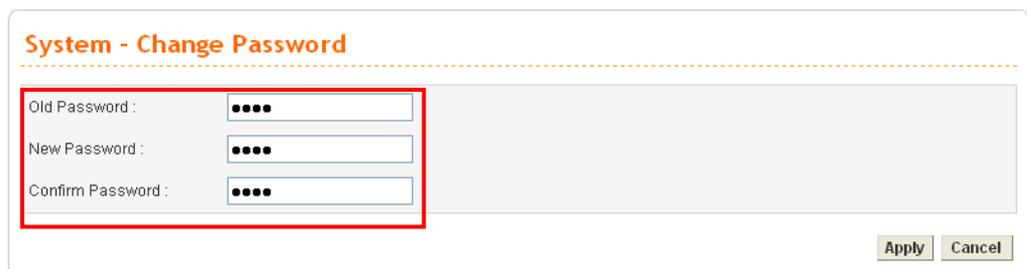
- Now, the **Main Screen** will pop up.



- Go to **System** page and choose **Change Password**.



- The following screen will appear.



- Enter the login password (1234) on the field of Old Password. Type a new one in the field of New Password and retype it on the field of Confirm Password. Then click **Apply** to continue.
- Now, the password has been changed. Next time, use the new password to access the Web Configurator for this adapter.

- Next, you will see the login screen after clicking **Apply**. Please use new password to re-enter the system configuration.



2.2 Quick Setup

Quick Setup is designed for configuring your broadband adapter accessing Internet with simply steps. There are two phases of quick setup, one is WAN configuration and the other is LAN configuration.

2.2.1 Adjusting WAN Connection Mode

In the **Quick Setup** group, you can configure the adapter to access the Internet with different modes such as Static and DHCP modes. For most users, Internet access is the primary application. The adapter supports the Ethernet WAN interface for Internet access. The following sections will explain in more detail the various broadband access configurations. All settings in this section will be applied in the first WAN1 interface.

Now, you have to select an appropriate WAN connection type for connecting to the Internet through this adapter according to the settings that your ISP provided.

MAC Address *Adapter Default-*

Use the default Mac address stored originally in adapter.

User Definition-

Use a MAC address defined by the user.

- Downstream Rate** Assign the downstream rate for this WAN interface. The default value is 102400 kbps (100 Megabit). This setting is very important for VigorTalk ATA-24 SH incoming buffer adjustment. If you use a DSL subscriber service with a 2Mbps downstream, please set the downstream rate setting with 2Mbps.
- Upstream Rate** Assign the transmission rate for this WAN interface. The default value is 102400 kbps (100 Megabit). This setting is very important for VigorTalk ATA-24 SH outgoing buffer adjustment. If you use a DSL subscriber service with a 256Kbps downstream, please set the downstream rate setting with 256Kbps.
- Type** Select a connection type for this WAN interface. Currently, there is only one setting offered for you to choose - Fast Ethernet.
- Physical Mode** Select connection speed mode for this WAN interface. There are **auto negotiation**, **full duplex**, and **half duplex** of either 10/100/1000M speed options for the WAN Interface.
- IP Mode** Select an IP mode for this WAN interface. There are two available modes for Internet access, **Static** or **DHCP**. On this page you may configure the WAN interface to use **Static** (fixed IP) or **DHCP** (dynamic IP address). Most of the cable users will use the **DHCP** mode to get a globally reachable IP address from the cable host system.

2.2.2 Static Mode

You can manually assign a static IP address to the WAN interface and complete the configuration by applying the settings and rebooting your adapter. Choosing **Static** as the IP mode, you will see the following page:

All the settings here are set by privately. Your ISP will not provide these settings.

- IP Address** Assign a private IP address to the WAN interface.
- Subnet Mask** Assign a subnet mask value to the WAN interface.
- Default Gateway** Assign a private IP address to the gateway.

Primary DNS	Assign a private IP address to the primary DNS.
Secondary DNS	Assign a private IP address to the secondary DNS.
IP Alias List	Assign other IP addresses to be bound to this interface. This setting is optional.

After setting up the **WAN** interface, the user can click **Next** to setup the **MGN** interface continuously.

MGN – MGN IP/DHCP Page

The screenshot shows the 'Quick Setup - MGN' configuration page. It is divided into two main sections: 'IP Configuration' and 'DHCP Server'.
IP Configuration:
 IP Address: 192.168.1.1
 Subnet Mask: 255.255.255.0
DHCP Server:
 Status: Enable Disable
 Start IP: 192.168.1.10
 End IP: 192.168.1.254
 Primary DNS: (empty field)
 Secondary DNS: (empty field)
 Lease Time (Min): 1440
 Gateway IP(Optional): (empty field)
 At the bottom right, there are two buttons: '<<Previous' and 'Finish'.

IP Address	Assign an IP address for the LAN interface.
Subnet Mask	Assign the subnet mask for the LAN interface.
Status	Click Enable to use DHCP server; click Disable to close DHCP server.
Start IP	Assign the start IP address of the IP pool that DHCP server can use for clients in LAN.
End IP	Assign the end IP address of the IP pool that DHCP sever can use for clients in LAN.
Primary DNS	Type the IP address for primary DNS.
Secondary DNS	Type the IP address for secondary DNS.
Lease Time	Type the number for lease time. The default setting is 1440.
Gateway IP	Type the IP address as DHCP client.

When you finished the above required settings, please click **Finish**. A system reboot page will appear. Click **Apply** to activate the static mode configuration.

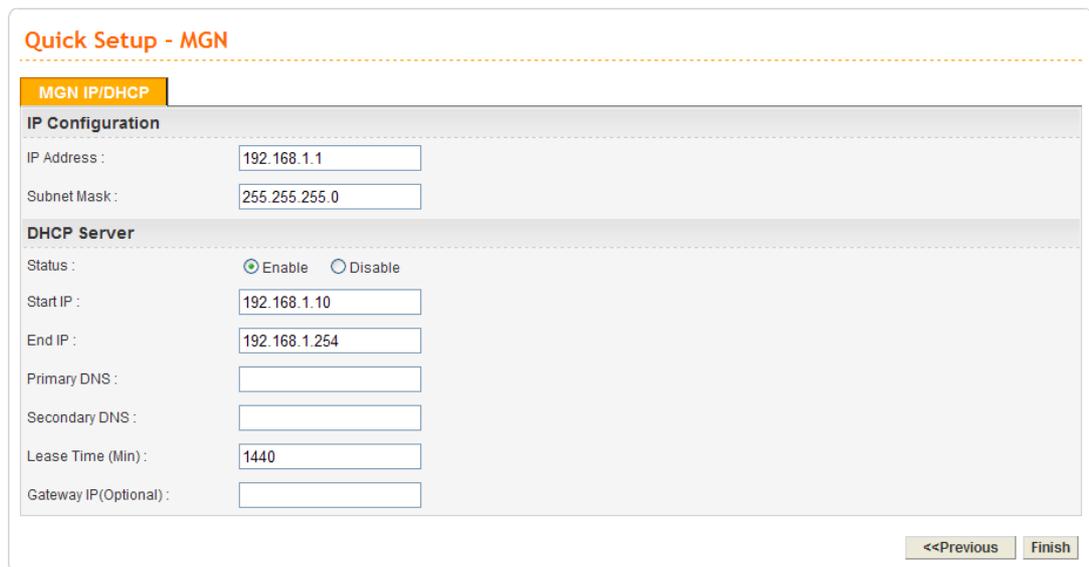
2.2.3 DHCP Mode

DHCP allows a user to obtain an IP address automatically from a DHCP server on the Internet. If you choose **DHCP** mode, the DHCP server of your ISP will assign a dynamic IP address for VigorTalk ATA-24 SH automatically. It is not necessary for you to assign any setting. (Host Name and Domain Name are required for some ISPs). Simply click **Next** to setup LAN interface.



After setting up the Host Name and Domain Name for the **WAN** interface, the user can click **Next** to setup the MGN interface continuously.

LAN – LAN IP/DHCP Page



- | | |
|--------------------|---|
| IP Address | Assign an IP address for the LAN interface. |
| Subnet Mask | Assign the subnet mask for the LAN interface. |
| Status | Click Enable to use DHCP server; click Disable to close DHCP server; click Relay Agent to activate relay agent function. |
| Start IP | Assign the start IP address of the IP pool that DHCP server can use for clients in LAN. |
| End IP | Assign the end IP address of the IP pool that DHCP sever can use for clients in LAN. |

Primary DNS	Type the IP address for primary DNS.
Secondary DNS	Type the IP address for secondary DNS.
Lease Time	Type the number for lease time. The default setting is 1440.
Gateway IP	Type the IP address as DHCP client.

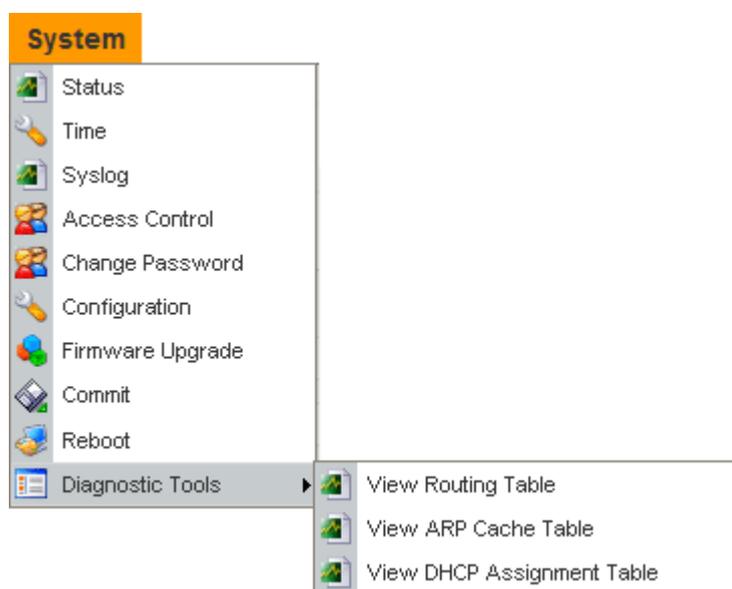
When you finished the above required settings, please click **Finish**. A system reboot page will appear. Click **Apply** to activate the static mode configuration.

3 Advanced Configuration

After finished basic configuration of the adapter, you can access Internet with ease. For the user who wants to adjust more setting for suiting his/her request, please refer to this chapter for getting detailed information about the advanced configuration of this adapter.

3.1 System setup

For the system setup, there are several items that you have to know the way of configuration: Status, Time Setup, Syslog Setup, Access Control Setup, Reboot and Firmware Upgrade Setup, Diagnostic Tools and Configuration Setup.



3.1.1 Status

The online **Status** function provides some useful system information on the current status of the VigorTalk ATA-24. A user can observe the system status on this Web page and determine which setting needed to be changed in corresponding web pages. In the **System** group, click the **Status** option. The online **Status** Web page contains three parts: **Basic Status, LAN Status, and WAN Status.**

Refresh Option You can choose to automatically refresh the Web page information. There are four options given as shown below.

No Refresh: Static information page.

Every 10 Seconds: Refreshes the page every 10 seconds.

Every 20 Seconds: Refreshes the page every 20 seconds.

Every 30 Seconds: Refreshes the page every 30 seconds.

Basic Status

General status of this adapter will be displayed on **Basic Status** page.

The screenshot shows the 'System - Status' page with a 'Refresh Option' dropdown set to 'No Refresh' and a 'Refresh' button. Below this are three tabs: 'Basic Status' (selected), 'MGN Status', and 'WAN Status'. The 'Basic Status' tab displays the following information:

Model :	ATA24 SH system
Hardware Version :	0
Firmware Version :	V2.1.3.1
Build Date&Time :	Mon Jul 20 11:35:12 CST 2009
System Uptime :	0 days 0 hours 16 minutes 45 seconds
CPU Usage :	0.0000%
Memory Size :	64 MBytes
Memory Usage :	59.9083%
Current System Time :	1970-01-01 00:16:45

Model	Displays the model name of the adapter.
Hardware Version	Displays the hardware version of the adapter.
Firmware Version	Displays the firmware version of the adapter.
Build Date&Time	Displays the date and time of the current firmware build.
System Uptime	Displays the amount of time that the adapter has been online.
CPU Usage	Displays the average percentage of the CPU being used.
Memory Usage	Displays the percentage of memory being used.
Current System Time	Displays the current local system time.

MGN Status

The status of MGN connection is shown in this page. Simply click **LAN Status** tag to get the detailed.

The screenshot shows the 'System - Status' page with the 'MGN Status' tab selected. The 'Refresh Option' dropdown is set to 'No Refresh' and there is a 'Refresh' button. The 'MGN Status' tab displays the following information:

IP Address :	192.168.1.1
MAC Address :	00:50:7f:c5:45:88
RX Packets :	608
TX Packets :	461

IP Address	Displays the IP address of the LAN interface.
MAC Address	Displays the MAC address of the LAN Interface.
RX Packets	Displays the total number of received packets at the LAN interface.

TX Packets Displays the total transmitted packets at the LAN interface.

WAN Status

The status of WAN interface (Static or DHCP) is shown in this page. Simply click **WAN Status** tag to get the detailed. There are four sets of WAN status can be shown in this page at one time. The sample below just lists one set of WAN status for only WAN1 interface is used.

The screenshot shows a web interface titled "System - Status". At the top, there is a "Refresh Option:" section with a dropdown menu set to "No Refresh" and a "Refresh" button. Below this is a navigation bar with three tabs: "Basic Status", "MGN Status", and "WAN Status" (which is highlighted in orange). The main content area is divided into two columns: "WAN1 :" and "WAN2 :". Each column lists various network parameters and their values.

WAN1 :		WAN2 :	
IP Address :	172.16.1.2	IP Address :	
MAC Address :	00:50:7f:c5:45:89	MAC Address :	00:50:7f:c5:45:8a
Primary DNS :	168.95.1.1	Primary DNS :	
Secondary DNS :	168.95.1.2	Secondary DNS :	
Gateway :	172.16.1.1	Gateway :	
RX Packets :	1909	RX Packets :	0
TX Packets :	153	TX Packets :	0
Connection Status :	disconnected	Connection Status :	
Up Time :		Up Time :	

IP Address Displays the IP address of the WAN interface.

MAC Address Displays the MAC address of the WAN Interface.

Primary DNS Displays the IP address of the primary DNS.

Secondary DNS Displays the IP address of the secondary DNS.

Gateway Displays the IP address of the default gateway.

RX Packets Displays the total received packets for each WAN interface.

TX Packets Displays the total transmitted packets for each WAN interface.

Connection Status Displays the connection status of the WAN interface.

Up Time Displays the total system uptime of the interface.

3.1.2 Time

As an NTP (Network Time Protocol) client, the adapter gets standard time from the time server. Typically, NTP achieves high accuracy and reliability with multiple redundant servers and diverse network paths.

The VigorTalk ATA-24 SH supports synchronization with a specific NTP server or the remote PC host of the administrator. In the **System** group, click the **Time** option. The Time page is shown below:

System - Time

Use Browser Time
 Use NTP Time

NTP Server :

Time Zone :

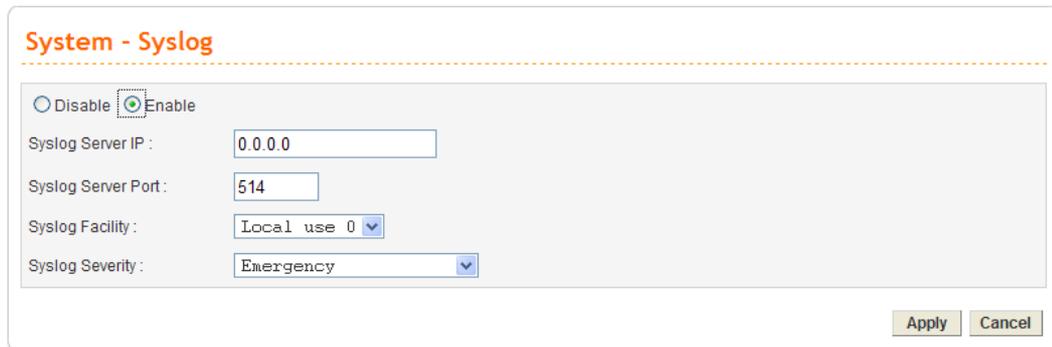
Daylight Saving Time : Not Use Use

Update Interval :

- Use Browser Time** Click this option to use the browser time from the remote administrator PC host as adapter's system time.
- Use NTP Time** Click this option to use the time from an NTP server as adapter's system time.
- NTP Server** Assign a public IP address or domain name of the NTP server.
- Time Zone** Select the time zone where the VigorTalk ATA-24 SH is located.
- Daylight Savings Time** Select **Use** to activate this function. This function is useful for some areas.
- Update Interval** Select a time interval for updating from the NTP server.
- Apply** Click **Apply** to save these settings.

3.1.3 Syslog

The VigorTalk ATA-24 SH supports a Syslog function to keep a record of abnormal conditions. The adapter will send Syslog packets to a Syslog server on the remote site. The administrator can observe any abnormal events from VigorTalk ATA-24. In the **System** group, click the **Syslog** option. The Syslog web page is shown below:



Status

Click **Enable** to activate this function. The adapter will send system log message for your reference. If you click **Disable**, the adapter will not send out any message about system log.

Syslog Server IP

The IP address of the Syslog server. If a user assigns an IP address of “0.0.0.0”, the Syslog function will be disabled. Then, VigorTalk ATA-24 SH will not send Syslog packets to the Syslog server.

Syslog Server Port

Assign a port for the Syslog protocol.

Syslog Facility

Facility value (local user 0 to 7) defined by syslog protocol. Use the drop down list to choose the one you want.

Syslog Severity

Determine what level of the log will be sent out. There are eight items representing different levels provided by the device.

Emergency: this is the highest level. Only Syslog marked with emergency will be sent out.

Alert: when you choose this level, Syslog marked with Emergency, Alert will be sent out.

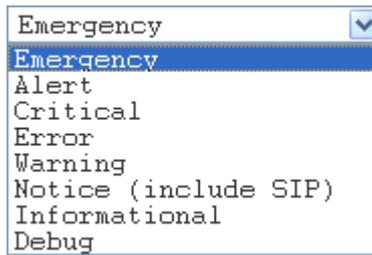
Critical: when you choose this level, Syslog marked with Emergency, Alert and Critical will be sent out.

Error: when you choose this level, Syslog marked with Emergency, Alert, Critical and Error will be sent out.

Warning: when you choose this level, Syslog marked with Emergency, Alert, Critical, Error and Warning will be sent out.

Informational: when you choose this level, Syslog marked with Emergency, Alert, Critical, Error, Warning and Information will be sent out.

Debug: when you choose this level, all of the logs will be sent out.



Apply

Click **Apply** to save these settings.

3.1.4 Access Control

This page allows you to determine which services (HTTP/Telnet/SSH) is used for the user to access VigorTalk ATA-24. In addition, you can also limit some hosts to access VigorTalk ATA-24 SH with specified IP address.

In the **System** group, click the **Access Control** option. You will get the following page:

System - Access Control

Management Method

Allow Management Method:

HTTP Telnet SSH

Management Access Control

Allow Management from the WAN

Disable Enable All Enable User Defined WAN IP

Allowed IP1: ~

Allowed IP2: ~

Allowed IP3: ~

Management Port

Default Ports (HTTP Port:80 Telnet Port:23 SSH Port:22) User Defined Ports

HTTP Port:

Telnet Port:

SSH Port:

PING Restriction

Disable PING from the LAN

Disable PING from the WAN

Management Method

There are three management methods provided here for you to choose for your adapter. Check HTTP/Telnet/SSH for the adapter.

Allow Management from the WAN

Disable - Disable the management from the WAN interface.

Enable All - Enable all management (through HTTP/Telnet/SSH) from the WAN interface.

Enable User Defined WAN IP - System can be managed by these three IP addresses via WAN.

Allowed IP1(to 3) - Type in ranges for IP addresses (up to three) for managing the system.

Management Port

Default Ports - Use the default ports for HTTP and Telnet if you choose HTTP and Telnet as management methods.

User Defined Ports - Or you can assign new port numbers for HTTP, Telnet and SSH respectively.

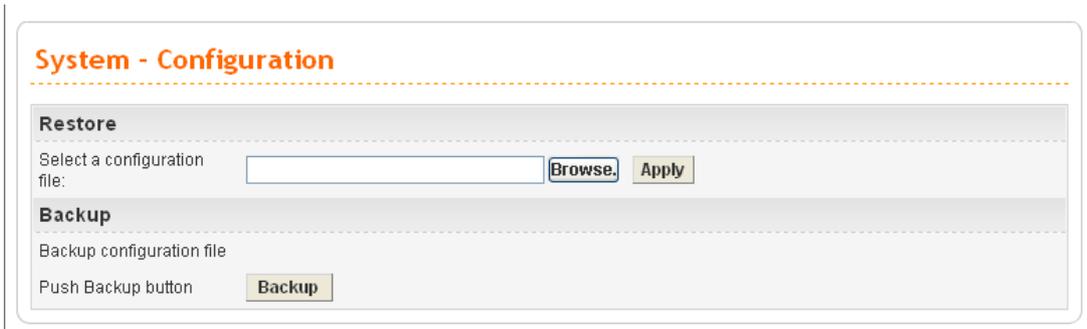
PING Restriction

Disable PING from the LAN -Choose this function to reject all ICMP packets from LAN side.

Disable PING from the WAN - Choose this function to reject all ICMP packets from WAN side.

3.1.5 Configuration

Most of the settings can be saved locally as a configuration file, and can be applied to another adapter. The VigorTalk ATA-24 SH supports the restoring and uploading functions of the **configuration files**. In the **System** group, click the **Configuration Setup** option. And you can see the following page.



Select a Configuration File Please click the **Browse...** button to find out the location of the configuration file to be uploaded to the adapter and click **Apply**.

Backup Configuration File Download the configuration file to a local host. The default file name is "ATA24-SH.cfg".

Push Backup Button

3.1.6 Firmware Upgrade

VigorTalk ATA-24 SH allows users to upgrade firmware through a Web interface. In the **System** group, click the **Firmware Upgrade** option. You can see the following page then. Before you execute the firmware upgrade, please download the **newest firmware** from Draytek's website (www.draytek.com) or FTP site ([ftp.draytek.com](ftp://ftp.draytek.com)) on the computer first.

System - Firmware Upgrade

Caution : After an upgrade procedure a reboot is required.

Current Version : ATA24 system V2.1.3.1

Location : Local Remote

Firmware :

TFTP Server IP

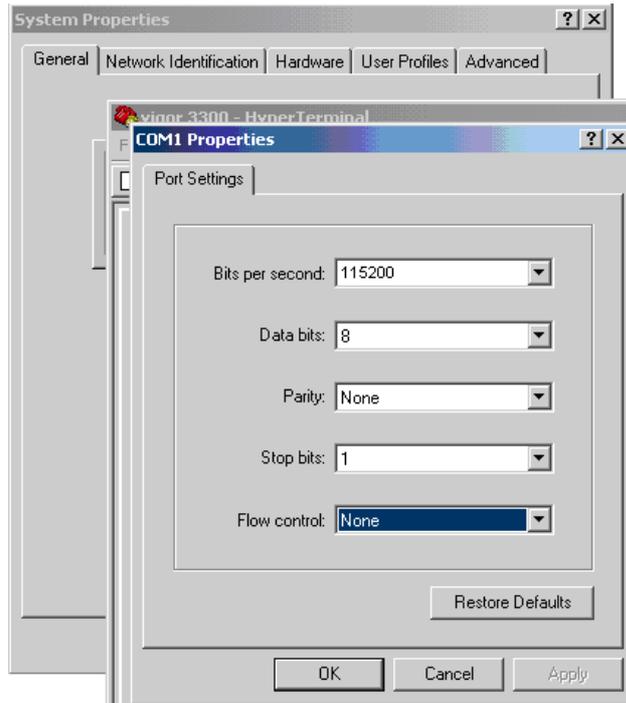
Remote File Name

Caution	Displays a caution for your reference.
Current Version	Displays current firmware version that you are using.
Location	<i>Local</i> means upgrade firmware from browser. <i>Remote</i> means upgrade firmware from a remote TFTP server.
Firmware	Specify the location of the firmware file if you want to upgrade the firmware locally.
TFTP Server IP	If you want to upgrade the firmware of this adapter from remote side, please type the IP address of the TFTP server.
Remote File Name	The default filename will be shown here. If you have use another name to save the firmware file, please type the new name in this field.
Apply	After finished your selection, please click Apply to execute the firmware upgrade.

Firmware Upgrade from a Console Port

Firmware upgrade can be done from a console port, too. The following example was run on a Windows environment.

1. Download the newest firmware from the DrayTek Website (www.draytek.com.tw) or FTP site ([ftp.draytek.com](ftp://ftp.draytek.com)) on your computer first.
2. Connect the RJ45 connector of console cable to the console port on VigorTalk ATA-24 SH and the DB9 connector of the console cable to the RS232 port on the PC.



The default setting of the console port is “baud rate 115200, no parity, and 8 bit with 1 stop bit.”

3. Power on VigorTalk ATA-24, then press **ENTER** before the system reboots completely.

```

*****
* ATA-24 Bootloader Version: V1.0.2 (Oct 3 2008 - 15:52:20) *
*****

Press [ENTER] key within 5 sec. to download image... 5

Current LAN IP is 172.17.3.102
New IP:

Current Serv IP is 172.17.3.234
New IP:

Current image is ata24_m825.all
New Name: █

```

4. Type LAN IP, TFTP Server IP, Image Name one by one, and press ENTER.
5. The firmware upgrade begins.
6. After firmware upgrade is finished, the device will restart.

```

Tera Term - COM4 VT
File Edit Setup Control Window Help
*****
* ATA-24 SH Boot loader Version: V1.0.5 (Jul 20 2009 - 11:20:16) *
*****
Press [ENTER] key within 5 sec. to download image... 0
.....
.....
.....
vlan4040: Setting MAC address to 00 50 7f c5 45 88.
device eth0 entered promiscuous mode
VLAN (vlan4040): Setting underlying device (eth0) to promiscuous mode.
vlan4041: Setting MAC address to 00 50 7f c5 45 89.
VLAN (vlan4041): Underlying device (eth0) has same MAC, not checking promiscuous mode.
vlan4042: Setting MAC address to 00 50 7f c5 45 8a.
vlan4043: Setting MAC address to 00 50 7f c5 45 8b.
dhcp: DHCP Server LAN0 initialize
Management interface started

Login: admin
Password:
ATA24-SH>
ATA24-SH>
ATA24-SH> interface vlan4041 up
Starting void:
Voip: initialize GOS module
Voip: initialize CFG module
voip: get ip info - wan_ip = 172.16.1.2, rtp_ip = 172.16.1.2
Voip: initialize CDR module
Voip: initialize STUN module (mode 0)
Voip: initialize SIP EM module
Voip: initialize SIP GL module
VoIP: Initialize CC Module ...
Voip: Initializing EPT Module ...
Voip: Initializing DSP Module ...
clear cache UNREACHABLE
into sigusr2_handler
Voip: Initializing CAS Module ...
Voip: Initialize LESLIC module
VoIP Ready
Voip: Initialize DWP SLIC After Boot
ATA24-SH>

```

3.1.7 Commit

Commit can save the current settings. Please click **Apply** to save the VigorTalk ATA-24 SH system settings.



3.1.8 Reboot

The VigorTalk ATA-24 SH system can be restarted from a Web browser. **Reboot** screen can appear after you finish the changing of WAN and LAN settings. You have to reboot the adapter to invoke the configured settings that you made before. Besides, you can select **Reset to factory default** to reboot the device and retrieve the default settings.

In the **System** group, choose the **Reboot** option. In the web page of **Reboot**, a user must either keep the current configuration settings or use the default configuration after the VigorTalk ATA-24 SH system has been rebooted.

System - Reboot

System rebooting will take 60 seconds

Reset to factory default

Apply

Click **Apply** to reboot the whole system. The rebooting procedure usually takes 70 or more seconds.

System is rebooting, please wait...

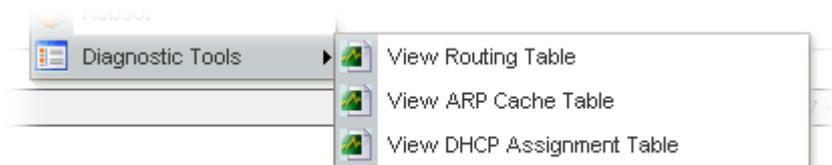
20 seconds left

If your current interface or management port configuration has been changed, please access with the new URL.

3.1.9 Diagnostic Tools

In some cases, a user may need to know some information about the adapter, such as static or dynamic databases, or other routing information. The VigorTalk ATA-24 SH supports four functions, **Routing Table**, **ARP Cache Table**, and **DHCP Assignment Table** for the user to review such information.

In the **System** group, click the **Diagnostic Tools** option



- Select **View Routing Table** to get the following page:

System - Diagnostic Tools - View Routing Table

Destination	Gateway	Subnet Mask	Flags	Interface
192.168.1.0	0.0.0.0	255.255.255.0	U	vlan4040
172.16.0.0	0.0.0.0	255.255.0.0	U	vlan4041
127.0.0.0	0.0.0.0	255.0.0.0	U	lo

Refresh

Destination Displays the destination IP address for various routings.

Gateway Displays the default gateway.

Subnet Mask Displays the subnet mask for various routings.

Flags Displays the status of the routing entries.

Interface Denoted by **vlan4040** if it is a LAN interface and **vlan4041** if it is a WAN interface.

Refresh Click **Refresh** to re-display this web page for getting newest routing information.

- Select **View ARP Cache Table** to get the following page:

Index	IP Address	MAC Address	Interface
1	172.16.3.8	00:1D:09:68:1D:8A	vlan4041
2	172.16.3.18	00:50:FC:2F:3D:17	vlan4041
3	192.168.1.10	00:0E:A6:2A:D5:A1	vlan4040
4	172.16.3.4	00:50:7F:C0:8D:84	vlan4041

IP Address Displays the IP address for different ARP cache.

MAC Address Displays the MAC address for different ARP cache.

Interface Denoted by **vlan4040** if it is a LAN interface. **vlan4041** means it is a WAN1 interface; **vlan4042** means it is a WAN2 interface; **vlan4043** means it is a WAN3 interface.

Refresh Click **Refresh** to re-display this web page for getting newest ARP information.

- Select **View DHCP Assignment Table** to get the following page:

Index	Assigned IP	MAC Address	Time Left
1	192.168.1.10	00:0E:A6:2A:D5:A1	23 hours, 52 minutes, 15 seconds

Assigned IP Displays the IP address of the static DHCP server.

MAC Address Displays the MAC address of the static DHCP server.

Time Left Displays the remaining time for this IP address assigned by DHCP server. When the time expired, such IP address would not be kept for this client and might be assigned to other client.

Refresh

Click **Refresh** to re-display this web page for getting newest routing information.

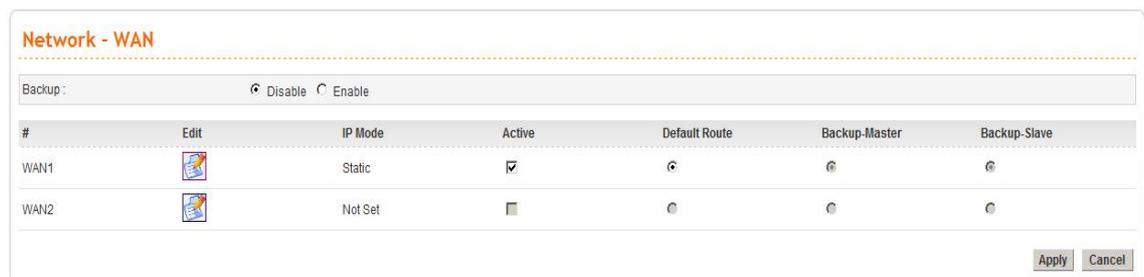
3.2 Network Setup

For Internet access, it is necessary for you to set **WAN** and **MGN** interfaces for the adapter.



3.2.1 WAN and Internet Access Setup

The VigorTalk ATA-24 SH supports two WAN interfaces (with two IP Modes – Static or DHCP), which share the same setting page. In the **Network** group, please click the **WAN** option. The following page will be shown.



Backup Enables or disables backup function for WAN interfaces. If you enable this function, the backup-master/backup-slave will execute the job of master/slave device when the master/slave device fails to work.

Edit Open the configuration page of this WAN interface.

IP Mode Displays current mode of this WAN interface. There are two options: Static or DHCP

Active Activates/closes this WAN interface.

Default Route Sets this WAN interface as default route interface.

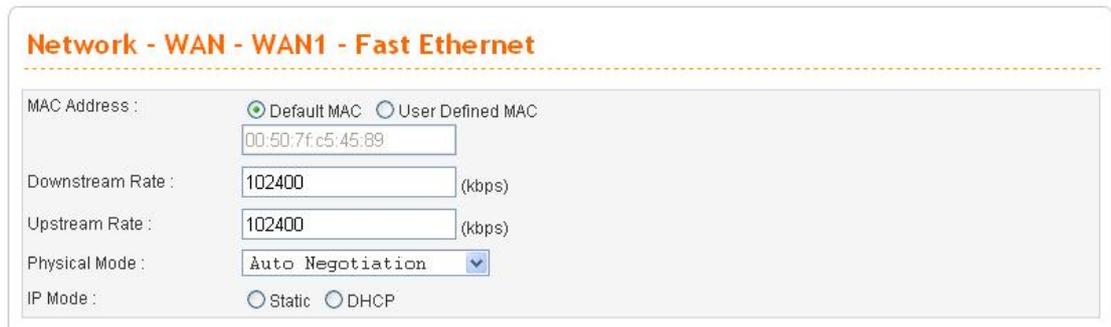
Backup-Master Sets this WAN interface as a master interface. WAN1 must be assigned as Master interface if Backup function is enabled.

Backup-Slave Sets this WAN interface as a slave interface.

VoIP Sets this WAN interface as VoIP default interface.

Most users will use their adapters primarily for Internet access. The VigorTalk ATA-24 SH supports broadband Internet access and provides multiple WAN interfaces. The following sections will give a detailed illustration to broadband access methods.

Click the “**Edit**” icon to bring up the WAN configuration page for the corresponding interface.



Network - WAN - WAN1 - Fast Ethernet

MAC Address : Default MAC User Defined MAC
00:50:7f:c5:45:89

Downstream Rate : (kbps)

Upstream Rate : (kbps)

Physical Mode : ▾

IP Mode : Static DHCP

Default MAC Uses the default Mac address.

User Defined MAC Uses a MAC address defined by users. If you select this item, you have to type the MAC address in the box below.

Downstream Rate Sets downstream rate for this WAN interface. The default value is 102400 kbps (100 Megabit).

Upstream Rate Sets transmission rate for this WAN interface. The default value is 102400 kbps (100 Megabit).

Physical Mode Sets connection speed mode. There are five options including **Auto negotiation, full duplex, half duplex, 10M, 100M** and **1000M**.

IP Mode Sets an IP Mode with **Static (fixed IP)** or **DHCP (dynamic IP address)** and creates the IP group information. Most cable modem users will use DHCP to get a globally reachable IP address from the cable head-end system. Different mode will lead different configuration and will be explained in later section.

Before you connect a broadband access device e.g. a DSL/Cable modem to VigorTalk ATA-24 SH, you need to know what kind of Internet access your ISP provides. The following sections introduce two widely used broadband access services: **Static** for DSL, **DHCP** for Cable modem. In most cases, you will get a DSL or cable modem from the broadband access service provider. VigorTalk ATA-24 SH is connected behind the broadband device i.e. DSL/cable modem and works as a NAT or IP adapter for broadband connections.

Next, we will introduce each WAN mode in detailed.

Static IP Configuration

It means that the IP group information for WAN interface is manually assigned by the user.

IP Mode : Static DHCP

Static/DHCP Configuration

IP Address : Host Name :

Subnet Mask : Domain Name :

Default Gateway : (Host Name and Domain Name are required for some ISPs.)

Primary DNS :

Secondary DNS :

MTU :

Connection Detection

Detect Type : ▼

Detect Interval(sec) :

No-Reply Count :

Detect Destination Host : (IP or Domain Name)

IP Alias List

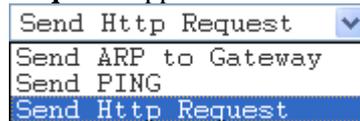
1.	<input type="text"/>	2.	<input type="text"/>
3.	<input type="text"/>	4.	<input type="text"/>
5.	<input type="text"/>	6.	<input type="text"/>
7.	<input type="text"/>	8.	<input type="text"/>

[9-32](#)

Apply Reset Cancel

IP Address	Sets the private IP address of WAN interface.
Subnet Mask	Sets the subnet mask value of WAN interface.
Default Gateway	Sets the private IP address of gateway.
Primary DNS	Sets the private IP address of primary DNS.
Secondary DNS	Sets the private IP address of secondary DNS.
MTU	It means the maximum transmission unit. Default value is 1500. Change it if you want.
Host Name	Some ISP may ask you to type your host name. Please type in if necessary.
Domain Name	Some ISP may ask you to type your domain name. Please type in if necessary.
Detect Type	Select a detecting type for this WAN interface. There are three ways Send ARP to Gateway , Send PING and Send HTTP

Request supported in ATA24-SH.



Detect Interval (sec)

Assign an interval period of time for each detecting. The minimum value is 3 and no limit for maximum value.

No-Reply Count

Assign detecting times to ensure the connection of the WAN. After passing the times you set in this field and no reply received by the adapter, the connection of WAN interface will be regarded as breaking down.

Detect Destination Host (IP or Domain Name)

Assign an IP address or Domain name as a destination to be detected whether the host is active (sending reply to the adapter) or not. If not, the connection of WAN interface will be regarded as breaking down. This function is available when **Detect Type** is set with **Send PING** or **Send Http Request**.

IP Alias List

Sets other IP addresses binding in this interface. You can set up to 32 sets of IP alias settings. If you have typed addresses here, you can see and choose it in later web page settings.

Apply

Click **Apply** to go back to the WAN Interface Configuration page. To apply all settings, click **Apply** on the WAN Interface Configuration page and reboot your adapter.

Reset

Click this button to clear all the configurations for this page.

DHCP Configuration

If the WAN interface is set as a DHCP client, the VigorTalk ATA-24 SH will ask for IP network settings from the DHCP server or DSL modem automatically. It is not necessary for users to manually configure the adapter.

IP Mode : Static DHCP

Static/DHCP Configuration

IP Address : Host Name :

Subnet Mask : Domain Name :

Default Gateway : (Host Name and Domain Name are required for some ISPs.)

Primary DNS :

Secondary DNS :

MTU :

Connection Detection

Detect Type : ▼

Detect Interval(sec) :

No-Reply Count :

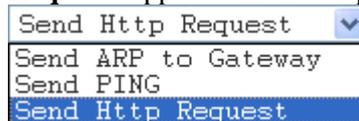
Detect Destination Host :
(IP or Domain Name)

MTU It means the maximum transmission unit. Default value is 1500. Change it if you want.

Host Name Some ISP may ask you to type your host name. Please type in if necessary.

Domain Name Some ISP may ask you to type your domain name. Please type in if necessary.

Detect Type Select a detecting type for this WAN interface. There are three ways **Send ARP to Gateway**, **Send PING** and **Send HTTP Request** supported in the adapter.



Detect Interval (sec) Assign an interval period of time for each detecting. The minimum value is 3 and no limit for maximum value.

No-Reply Count Assign detecting times to ensure the connection of the WAN. After passing the times you set in this field and no reply received by the adapter, the connection of WAN interface will be regarded as breaking down.

Detect Destination Host (IP or Domain Name) Assign an IP address or Domain name as a destination to be detected whether the host is active (sending reply to the adapter) or not. If not, the connection of WAN interface will be regarded as breaking down. This function is available when **Detect Type** is set with **Send PING** or **Send Http Request**.

Apply

Click **Apply** to go back to the WAN Interface Configuration page. To apply all settings, click **Apply** on the WAN Interface Configuration page and reboot your adapter.

Reset

Click this button to clear all the configurations for this page.

3.2.2 MGN

In the **Network** group, select **MGN** option. The following page for LAN IP/DHCP will be shown.

The screenshot shows a web interface for configuring network settings. The title is "Network - MGN". Below the title is a tab labeled "MGN IP/DHCP". The interface is divided into two main sections: "IP Configuration" and "DHCP Server".

IP Configuration:

- IP Address : 192.168.1.1
- Subnet Mask : 255.255.255.0

DHCP Server:

- Status : Enable Disable
- Start IP : 192.168.1.10
- End IP : 192.168.1.254
- Primary DNS : [Empty field]
- Secondary DNS : [Empty field]
- Lease Time (Min) : 1440
- Gateway IP(Optional) : [Empty field]

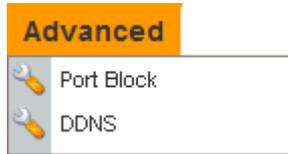
At the bottom right of the form, there are two buttons: "Apply" and "Cancel".

- | | |
|------------------------------|---|
| IP Address | Type the IP address for LAN/DHCP. |
| Subnet Mask | Type the subnet mask for the LAN IP/DHCP. |
| Status | Click Enable the DHCP server; click Disable to close DHCP server. |
| Start IP | Sets the starting IP address of the IP address pool for DHCP server. |
| End IP | Sets the ending IP address of the IP address pool for DHCP server. |
| Primary DNS | Sets the private IP address of the primary DNS. |
| Secondary DNS | Sets the private IP address of the secondary DNS. |
| Lease Time (Min) | Sets a lease time for the DHCP server. The time unit is minute. |
| Gateway IP (Optional) | Sets a gateway IP address for the DHCP server. |

Click **Apply** to reboot the system and apply the settings.

3.3 Advanced Setup

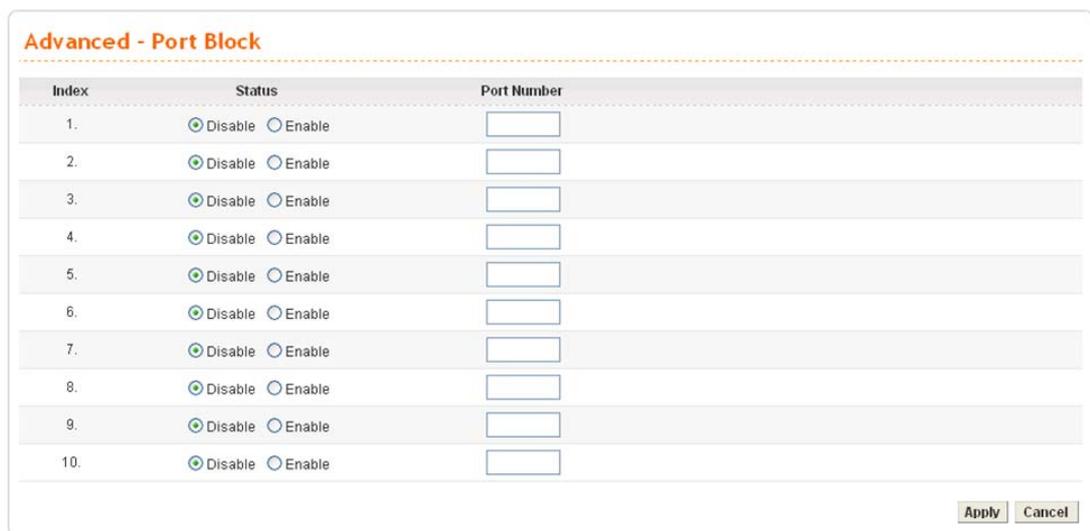
In the **Advanced** menu, there are several items offered here for you to adjust for the adapter.



3.3.1 Port Block

The **Port Block** function provides a user to set lots of proprietary port numbers. Packets will be dropped if destination ports (both TCP and UCP) of packets with these assigned port numbers are on WAN and LAN. The advantage of this feature is to filter some unnecessary packets or attacking packets on Internet environment or LAN network. VigorTalk ATA-24 SH supports ten port numbers to be blocked.

In the **Advanced** group, click **Port Block** option. You will get the following page.

A screenshot of the 'Advanced - Port Block' configuration page. The page has a title bar 'Advanced - Port Block' in orange. Below the title bar is a table with three columns: 'Index', 'Status', and 'Port Number'. The table has 10 rows, each representing a port to be blocked. Each row has a radio button for 'Disable' (which is selected) and a radio button for 'Enable'. To the right of each row is a text input field for the port number. At the bottom right of the page, there are two buttons: 'Apply' and 'Cancel'.

Index

The number of each entry.

Status

User can **Disable** or **Enable** the port block for the specified port.

Port Number

Assign a port number to be blocked in system. The default port setting is 135.

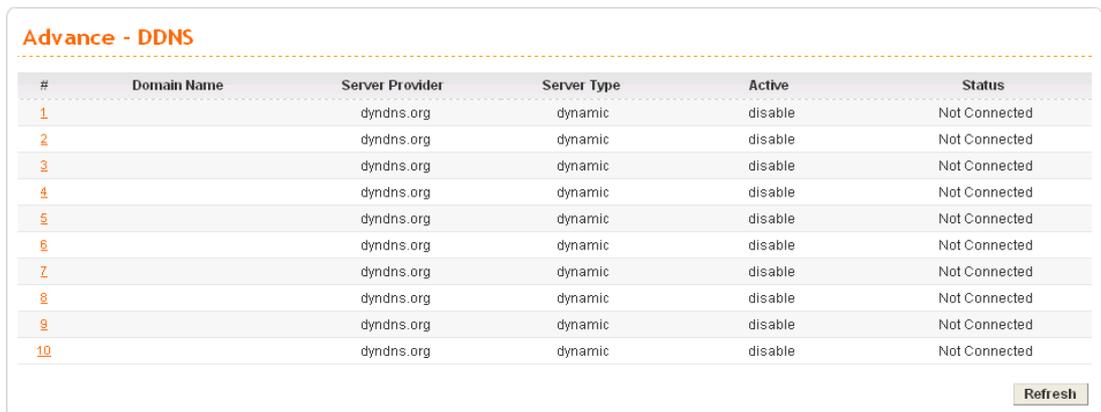
Click **Apply** to finish this setting.

3.3.2 DDNS

The Dynamic DNS function allows the adapter to update its online WAN IP address, which assigned by ISP or other DHCP server to the specified Dynamic DNS server. Once the adapter is online, you will be able to use the registered domain name to access the adapter or internal virtual servers from the Internet. DDNS is more popular on dynamic IP users, who typically receive dynamic, frequently-changing IP addresses from their service provider.

Before you set up the Dynamic DNS function, you have to subscribe free domain names from the Dynamic DNS service providers. The adapter provides up to ten accounts for the function and supports the following providers: **www.dynsns.org**, **www.no-ip.com**, **www.dtdns.com**, **www.changeip.com**, **www.ddns.cn**. You should visit their websites for registering your own domain name on the adapter.

In the **Advanced** group, click **DDNS** option. You will get the following page.



The screenshot shows a web interface titled "Advance - DDNS". It contains a table with the following columns: #, Domain Name, Server Provider, Server Type, Active, and Status. There are 10 rows of data, all with "dyndns.org" as the server provider and "dynamic" as the server type. The "Active" column shows "disable" for all entries, and the "Status" column shows "Not Connected". A "Refresh" button is located at the bottom right of the table.

#	Domain Name	Server Provider	Server Type	Active	Status
1		dyndns.org	dynamic	disable	Not Connected
2		dyndns.org	dynamic	disable	Not Connected
3		dyndns.org	dynamic	disable	Not Connected
4		dyndns.org	dynamic	disable	Not Connected
5		dyndns.org	dynamic	disable	Not Connected
6		dyndns.org	dynamic	disable	Not Connected
7		dyndns.org	dynamic	disable	Not Connected
8		dyndns.org	dynamic	disable	Not Connected
9		dyndns.org	dynamic	disable	Not Connected
10		dyndns.org	dynamic	disable	Not Connected

- Domain Name** Display the domain name set for the entry.
- Service Provider** Display the service provider that supports DDNS.
- Service Type** Display the service type for the entry.
- Active** Display the activation status (disable or enable) for this entry.
- Status** Display the connection status of this entry.

Click **Refresh** to re-display the whole page information.

To modify DDNS setting, click an entry number to get into edit mode.

Advanced - DDNS Setting

Status : Disable Enable

Interface :

Server Provider :

Server Type :

Domain Name :

Login Name :

Login Password :

Wild Card : Disable Enable

Backup MX : Disable Enable

Mail Extender :

Status Click **Disable** to disable this function. Click **Enable** to activate this function.

Interface Select a specific interface for registering on DDNS server. The Interface should be any WAN port on VigorTalk series.

Server Provider Assign a provider name to support DDNS server. The VigorTalk ATA-24 SH supports 7 domain server providers as default.

-
-
-
-
-
-
-

Server Type Select **Static**, **Dynamic** or **Custom** type for this entry of DDNS settings.

Domain Name Assign a private domain name to be accessed.

Login Name Assign a name to login into DDNS server.

Login Password Assign a password to login into DDNS server.

Wild Card If you want anything-here.yourhost.dyndns.org to work (EX. To make things like www.yourhost.dyndns.org work), click “Enable” to active this function.

Backup MX MX stands for Mail Exchanger. Mail Exchangers are used for directing mail to specific servers other than the one a hostname points at.

Mail Extender Assign an email address.

Click **Apply** to finish these settings and return to previous page.

Note:

1. The Wildcard and Backup MX features are not supported for all Dynamic DNS providers. You could get more detailed information from their websites.
2. Backup MX provides a secondary mail server to hold your e-mail if your main email server go offline for any reason. Once you go back online, your email will be delivered to you.

3.4 Firewall Setup

The firewall controls the allowance and denial of packets through the adapter. The **Firewall Setup** in the VigorTalk ATA-24 SH Series mainly consists of Denial of Service (DoS) only. The firewall filters help to protect your computer against attack from outsiders.

The following sections will explain how to configure the **Firewall**. The **DoS** facility can detect and mitigate the DoS attacks. T



3.4.1 DoS

The DoS function helps to detect and mitigates DoS attacks. These include flooding-type attacks and vulnerability attacks. Flooding-type attacks attempt to use up all your system's resources while vulnerability attacks try to paralyze the system by offending the vulnerabilities of the protocol or operation system.

In the **Firewall** group, click the **DOS** option. You will see the following page. The DoS Defense Engine inspects each incoming packet against the attack signature database. Any packet that may paralyze the host in the security zone is blocked. The DoS Defense Engine also monitors traffic behavior. Any anomalous situation violating the DoS configuration is reported and the attack is mitigated.

A screenshot of the 'Firewall - DoS' configuration page. At the top, it says 'Firewall - DoS' in orange. Below that, there are radio buttons for 'DoS Defense' with 'Disable' selected and 'Enable' selected. The 'Enable' option is highlighted with a dashed box. Below this are several checkboxes for enabling various defenses: SYN flood, UDP flood, ICMP flood, Port Scan detection, Block IP options, Block Land, Block Smurf, Block trace route, Block SYN fragment, Block Fraggle Attack, Block TCP flag scan, Block Tear Drop, Block Ping of Death, Block ICMP fragment, and Block Unknown Protocol. Each of the first three flood defense options has a 'Threshold' and 'Timeout' field. The thresholds are all set to 300 Packets/sec and the timeouts are all set to 10 sec. At the bottom right, there are 'Apply' and 'Cancel' buttons.

DoS Defense	Enables or disables the DoS Defense function. The default value is Disable .
Enable SYN Flood Defense	Activates the SYN flood defense function. If the amount of TCP SYN packets from the Internet exceeds the user-defined threshold value, the adapter will be forced to randomly discard the subsequent TCP SYN packets within the user-defined timeout period. The default setting for threshold and timeout are 300 packets per second and 10 seconds, respectively.
Enable UDP Flood Defense	Activates the UDP flood defense function. If the amount of UDP packets from the Internet exceeds the user-defined threshold value, the adapter will be forced to randomly discard the subsequent UDP packets within the user-defined timeout period. The default setting for threshold and timeout are 300 packets per second and 10 seconds, respectively.
Enable ICMP Flood Defense	Activates the ICMP flood defense function. If the amount of ICMP echo requests from the Internet exceeds the user-defined threshold value, the adapter will discard the subsequent echo requests within the user-defined timeout period. The default setting for threshold and timeout are 300 packets per second and 10 seconds, respectively.
Enable Port Scan Detection	Activates the Port Scan detection function. Port scan sends packets with different port numbers to find available services, which respond. The adapter will identify it and report a warning message if the port scanning rate in packets per second exceeds the user-defined threshold value. The default threshold is 300 pps (packets per second).
Enable Block IP Options	Activates the Block IP options function. The adapter will ignore any IP packets with IP option field appearing in the datagram header.
Enable Block Land	Activates the Block Land function. A Land attack occurs when an attacker sends spoofed SYN packets with identical source address, destination addresses and port number as those of the victim.
Enable Block Smurf	Activates the Block Smurf function. The adapter will reject any ICMP echo request destined for the broadcast address.
Enable Block Trace Route	Activates the Block trace route function. The adapter will not forward any trace route packets.
Enable Block SYN Fragment	Activates the Block SYN fragment function. Any packets having the SYN flag and fragmented bit sets will be dropped.

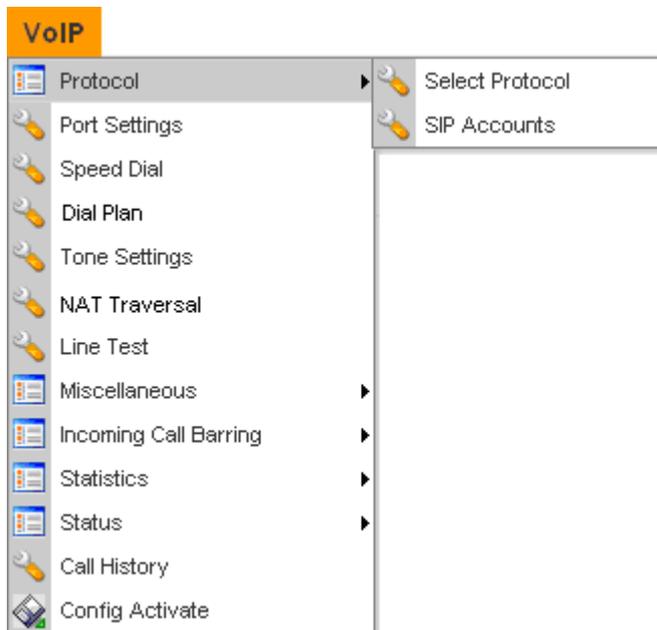
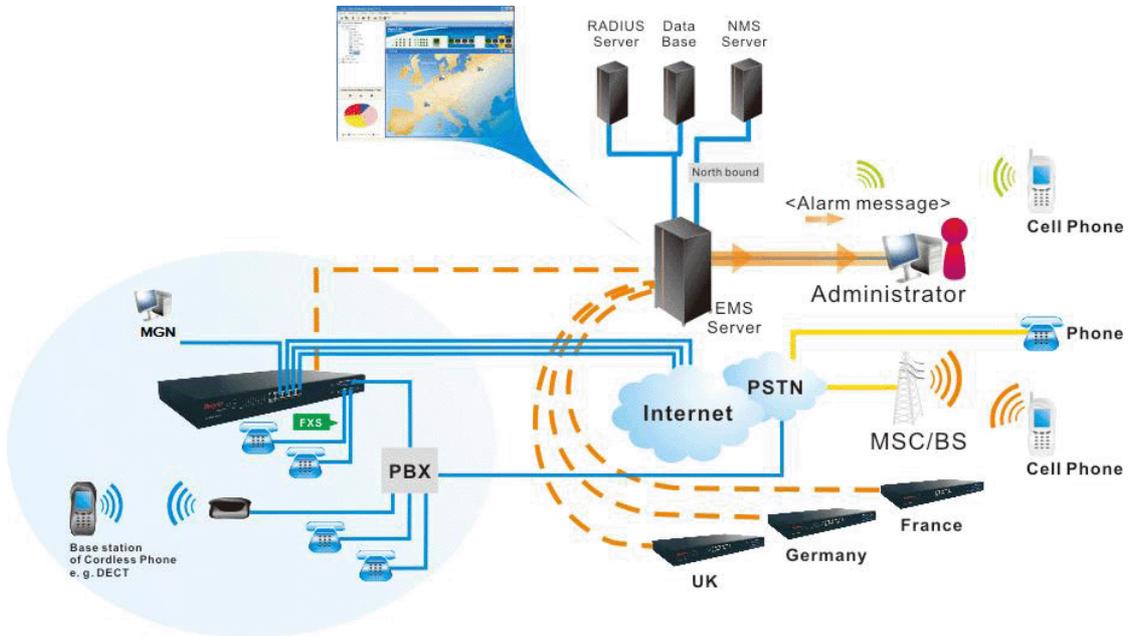
Enable Block Fraggle Attack	Activates the Block fraggle Attack function. Any broadcast UDP packets received from the Internet are blocked.
Enable TCP Flag Scan	Activates the Block TCP flag scan function. Any TCP packet with an anomalous flag setting is dropped. These scanning activities include no flag scan, FIN without ACK scan, SYN FIN scan, Xmas scan and full Xmas scan .
Enable Tear Drop	Activates the Block Tear Drop function. This attack involves the perpetrator sending overlapping packets to the target hosts so that target host will hang once they re-construct the packets. The adapters will block any packets resembling this attacking activity.
Enable Ping of Death	Activates the Block Ping of Death function. Many machines may crash when receiving an ICMP datagram that exceeds the maximum length. The adapter will block any fragmented ICMP packets with a length greater than 1024 octets.
Enable Block ICMP Fragment	Activates the Block ICMP fragment function. Any ICMP packets with fragmented bit sets are dropped.
Enable Block Unknown Protocol	Activates the Block Unknown Protocol function. The adapter will block any packets with unknown protocol types.

Click **Apply** to apply the settings when you finish the configuration.

3.5 VoIP Setup

Voice over Internet Protocol (VoIP) is a technology that allows you to make telephone calls using a broadband Internet connection instead of a regular (or analog) phone line.

The VigorTalk ATA-24 SH series provides cost effective voice solution for SME customers which can be explained with the following diagram.



3.5.1 Protocol

Select Protocol

There are three protocols can be used for VoIP phones – SIP, MGCP and H248. You should click either one of buttons to set corresponding settings for VoIP phones. Be aware that both sides (local end and remote end) should use same protocol for VoIP phones.

VoIP - Protocol

Select Protocol : SIP MGCP H248

SIP Server Configuration

MGCP Configuration

H248 Configuration

SIP Local Port :

#	Active	Outbound Proxy	Proxy Name	Proxy Address	Proxy Port	Registrar Addr	Registrar Port	Expires (sec)	Domain
1.	<input type="checkbox"/>	<input type="checkbox"/>	<input type="text"/>	<input type="text" value="0"/>	<input type="text" value="5060"/>	<input type="text" value="0"/>	<input type="text" value="5060"/>	<input type="text" value="3600"/>	<input type="text" value="0"/>
2.	<input type="checkbox"/>	<input type="checkbox"/>	<input type="text"/>	<input type="text" value="0"/>	<input type="text" value="5060"/>	<input type="text" value="0"/>	<input type="text" value="5060"/>	<input type="text" value="3600"/>	<input type="text" value="0"/>
3.	<input type="checkbox"/>	<input type="checkbox"/>	<input type="text"/>	<input type="text" value="0"/>	<input type="text" value="5060"/>	<input type="text" value="0"/>	<input type="text" value="5060"/>	<input type="text" value="3600"/>	<input type="text" value="0"/>

Example iptel iptel.org iptel.org iptel.org

Proxy **User-Agent Name**

1.	<input type="text" value="DrayTek ATA24-1.1.0"/>
2.	<input type="text" value="DrayTek ATA24-1.1.0"/>
3.	<input type="text" value="DrayTek ATA24-1.1.0"/>

● For SIP Configuration

- SIP Local Port** Type the port number for SIP protocol. The default value is 5060.
- Active** Click this box to activate this SIP proxy server setting.
- Outbound Proxy** Check this box to enable this function for sending SIP protocol packets to an SIP proxy server.
- Proxy Name** Type the name of the SIP proxy server.
- Proxy Address** Type the IP address of the SIP proxy server.
- Proxy Port** Type the port number of the SIP proxy server.
- Registrar Address** Type the IP address or domain name of the SIP registrar server.
- Registrar Port** Type the port number of the SIP registrar server.
- Expires** Type the timeout value for SIP protocols. The default value is 300.
- Domain** Type the IP address or domain name of the SIP Domain/Realm.
- User-Agent Name** Type the name for the client’s device.

You can set up to 3 sets of SIP configurations in this page.

● For MGCP Configuration

VoIP - Protocol

Select Protocol : SIP MGCP H248

SIP Server Configuration | **MGCP Configuration** | H248 Configuration

CallAgent 1

MGCP Call Agent Address :

MGCP Call Agent Port :

CallAgent 2

MGCP Call Agent Address :

MGCP Call Agent Port :

General Setting

MGCP Local Port :

EndPoint Name Style : aaln#[ip_addr] mac_addr#[ip_addr] aaln#[mac_addr]

aaln#[]

Logic ID Starting Number : (Range:1 ~ 99999)

Wild-carded RSIP : Each endpoint sends its own RSIP Send only one wild RSIP

Range Wildcard RSIP : Disable Enable

HeartBeat : Disable Enable

HeartBeat Period : (Range:1 ~ 65535)

HeartBeat Retry : (Range:1 ~ 300)

MGCP Call Agent Address The IP address of the Call Agent server in MGCP.

MGCP Call Agent Port The UDP port number for the Call Agent server.

MGCP Local Port The UDP port number in MGCP local terminal.

EndPoint Name Style Choose a proper name style for the VoIP settings. There are three options for you to choose.

aaln#[ip_addr] - ex: aaln/1@[1.1.1.1]

mac_addr#[ip_addr]- ex: 000504030201/1@[1.1.1.1]

aaln#[mac_addr]- ex: aaln/1@000504030201

aaln#[@] - ex: aaln/1@v3300.draytek.com

Logic ID Starting Number The starting number for “#” used in EndPoint Name Style. The range for the number is from 1 to 24. That is, if you type 3 in this field, the number 3 to 26 will be available for applying

Wild-carded RSIP For VoIP phone call with MGCP configuration, each port will send RSIP to call agent for notifying that port is initiated or restarted.

Each endpoint sends its own RSIP – Each port must send one RSIP message (e.g., aaln/1@[172.16.3.5]) to call agent respectively.

Send only one wild RSIP – Only one RSIP message (e.g., aaln/*@[172.16.3.5]) will be sent to call agent to indicate all ports are initiated/restarted.

Range Wildcard RSIP

Click **Enable** to send out RSIP message (e.g., aaln/*@[172.16.3.5]).

Click **Disable** to close such function.

HearBeat

Click **Enable** to check if MGCP server can work normally or not, otherwise click **Disable**.

HearBeat Period

Type the interval for the system to check the MGCP server.

HearBeat Retry

Type the times for the system to check the MGCP server.

● For H248 Configuration

VoIP - Protocol

Select Protocol : SIP MGCP H248

SIP Server Configuration | MGCP Configuration | **H248 Configuration**

CallAgent 1

H248 Call Agent Address : 172.16.1.100

H248 Call Agent Port : 2944

CallAgent 2

H248 Call Agent Address : 172.16.2.100

H248 Call Agent Port : 2944

Message ID

H248 MessageID Mode : [IPAddress]:Port [IPAddress] Manual message ID <00507R000000> 2944

H248 MessageID Address : 210.21.21.11

H248 MessageID Port : 2944

General Setting

H248 Local Port : 2944

H248 Termination Register : Disable Enable

SoftSwitch Type : Normal

HeartBeat : Disable Enable

HeartBeat Period : 60 (Range:30 ~ 65535)

HeartBeat Retry : 3 (Range:1 ~ 300)

H248 DigitMap

Short Timer : 3

Long Timer : 5

Start Timer : 20

Codec

Prefer Codec : G729A

Use Prefer Codec Only : Disable Enable

RFC 2833 DTMF event : Disable Enable

Default payload type for RFC 2833 : 101

Apply Cancel

H248 Call Agent Address

The IP address of the Call Agent server in H248.

H248 Call Agent

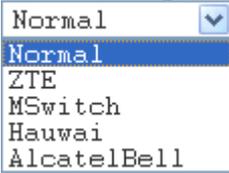
The port number for the Call Agent server in H248.

H248 MessageID Mode

Choose one of the modes for MessageID (defined by H248). Settings configured in **Message ID** field are used to be identified by the server supported with H248.

H248 MessageID IP Address

If you choose **IPAddress:Port** or **IPAddress** as the MessageID Mode, you have to type IP address manually in this field.

H248 MessageID Port	The port number for MessageID.
H248 Local Port	The UDP port number in H248 local terminal.
H248 Termination Register	Register for the port respectively. Disable – Close such function. Enable – If such device has been registered successfully, each port will register again after clicking Enable .
SoftSwitch Type	Use the drop down list to choose the manufacturer of the server you have to avoid the problem of IOP. If you have no idea of selection, simply choose Normal . 
HearBeat	Click Enable to check if H248 server can work normally or not, otherwise click Disable .
HearBeat Period	Type the interval for the system to check the H248 server.
HearBeat Retry	Type the times for the system to check the H248 server.
Short Timer	It is used to set the default H248 digitmap short timer. This timer is usually provided by the call agent. If it is not specified by the call agent, the default value will be applied.
Long Timer	It is used to set the default H248 digitmap long timer. This timer is usually provided by the call agent. If it is not specified by the call agent, the default value will be applied.
Start Timer	It is used to set the default H248 digitmap start timer. This timer is usually provided by the call agent. If it is not specified by the call agent, the default value will be applied.
Prefer Codec	Assign the prefer RTP codec used by H248 protocol. There are four possible values, G.711-ULAM, G.711-ALAW, G.723, and G.729A.
User Prefer Code Only	Disable – Disable this function. Enable - Enable this option to make H248 protocol using the Prefer Codec only. Only preferred codec will be used for outgoing and incoming calls. If the remote end does not support such Codec, the VoIP communication will be failed.
RFC 2833 DTMF event	Disable - Disable the RFC2833 (outband) DTMF event for H248 Protocol. Vigor device will send the DTMF tone as audio directly when you press the keypad on the phone (Use inband signal for DTMF event). Enable –Enable the RFC2833 (outband) DTMF event for H248

Protocol. Vigor device will capture the keypad number you pressed and transform it into digital form for sending to the other side. The receiver will generate the tone according to the digital form it received. This function is very useful when the network traffic congestion occurs, and also it can remain the accuracy of DTMF tone.

Default payload type for RFC 2833 The payload type value will be shown on RFC2833 packet header. This value is negotiated by the H248 client and server. This option set the default value for RFC2833 payload Type.

SIP Accounts

You have to set up your own SIP settings. When you apply for an account, your SIP service provider will give you relational information for you to type in this page.

VoIP - SIP Accounts						
#	User Name	Proxy Server	VoIP IP Address	Ring Port	Ring Type	Call Forwarding
1	<input checked="" type="radio"/> 1001		WAN	1	All Ports	
2	<input type="radio"/> 1002		WAN	2	All Ports	
3	<input type="radio"/> 1003		WAN	3	All Ports	
4	<input type="radio"/> 1004		WAN	4	All Ports	
5	<input type="radio"/> 1005		WAN	5	All Ports	
6	<input type="radio"/> 1006		WAN	6	All Ports	
7	<input type="radio"/> 1007		WAN	7	All Ports	
8	<input type="radio"/> 1008		WAN	8	All Ports	

1 2 3 4

- User Name** 1001 ~ 1032 are the default name specified by the system. Please click Edit to modify it if necessary.
- Proxy Server** Display the name of Proxy Server specified for such account.
- VoIP IP Address** Display the interface for such account. Such interface is used to apply VoIP traffics.
- Ring Port** Display the ring port number when the specified SIP account rings.
- Ring Type** Display the ring type for all the phone ports.
All Ports – Phones of all ports in the same group will ring.
First Available – The unconnected or unused phone of the first port will ring.
Round Robin – Only one phone will ring at one time and all of the phones will ring in turn.
- Call Forwarding** Display results for call forwarding.
Blank - no call forwarding.
Always - call forwarding for all of the calls.
Busy – call forwarding while the phone is busy.

No Answer - the call forwarding phone rings three times and gets no answer.

To edit an SIP account, please choose one of the radio buttons under Username and click Edit. The following page will be shown automatically.

VoIP - SIP Accounts - Edit

1

Disable Enable

Username:

Password:

Display Name:

Authentication ID:

Proxy Server:

Call without Registration: Disable Enable

Call Forwarding

Disable

Callforwarding all calls

Callforwarding busy

Callforwarding no answer after rings (Range:1~10)

SIP URL (Example:8001@iptel.org)

Subscribe for MWI

Disable Enable

Expires time:

MWI Inform

Play Special Dial Tone

Light up the phone lamp

CLIP

IncomingCall CLIP display:

OutgoingCall CLIP hidden:

Call Park

Call Park Dial Number:

IncomingCall Rings

Rings all ports in the group Rings the first available port Rings by round robin

Ring Port Setting

<input checked="" type="checkbox"/> P1	<input type="checkbox"/> P2	<input type="checkbox"/> P3	<input type="checkbox"/> P4	<input type="checkbox"/> P5	<input type="checkbox"/> P6	<input type="checkbox"/> P7	<input type="checkbox"/> P8
<input type="checkbox"/> P9	<input type="checkbox"/> P10	<input type="checkbox"/> P11	<input type="checkbox"/> P12	<input type="checkbox"/> P13	<input type="checkbox"/> P14	<input type="checkbox"/> P15	<input type="checkbox"/> P16
<input type="checkbox"/> P17	<input type="checkbox"/> P18	<input type="checkbox"/> P19	<input type="checkbox"/> P20	<input type="checkbox"/> P21	<input type="checkbox"/> P22	<input type="checkbox"/> P23	<input type="checkbox"/> P24

Disable/Enable

Click **Disable** to close this setting. Click **Enable** to activate this setting.

Username

Enter your account name of SIP Address, e.g. every text before @.

Password

The password provided to you when you registered with a SIP service.

Display Name

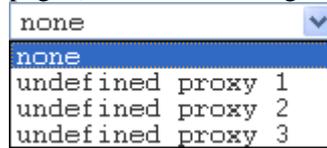
The caller-ID that you want to be displayed on your friend's screen.

Authentication ID

Type the name or number used for SIP Authorization with SIP Registrar.

Proxy Server

Before you choose, please set SIP proxy server first in previous page (SIP Server Configuration).

**Call without Registration**

Some SIP server allows user to use VoIP function without registering. For such server, please click **Enable** to invoke **Call without register**.

VoIP Address

The interface is used to apply VoIP traffics. There are two options: **WAN** and **LAN/VPN**. If LAN/VPN is selected, VoIP can be applied through a VPN tunnel to create a high security voice phone.

**Call Forwarding**

There are four options for you to choose.

Disable- It is to close call forwarding function.

Callforwarding busy- It means the incoming calls will be forwarded into SIP URL only when the local system is busy.

Callforwarding no answer after- It means if the incoming calls do not receive any response, they will be forwarded to the SIP URL by the time out.

SIP URL- Type in the SIP URL (e.g., aaa@draytel.org or abc@iptel.org) as the site for call forwarded.

Subscribe for MWI

This function is used to set SIP account for sending a message to the proxy server for subscribing MWI (Message Waiting Indicator). Part proxy server may need such subscription, yet not all of the proxy servers need.

Disable – Nothing will be sent out.

Enable - The device will send the information of Subscriber to the Server periodically.

Expires time – Set the time of expiring if you enable this function.

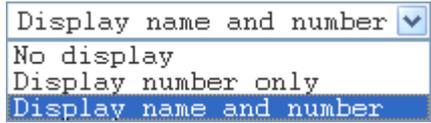
MWI Inform

Play Special Dial Tone – Play congest tone for five seconds while off-hook to inform you MWI message.

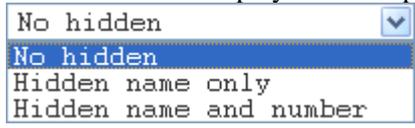
Light up the phone lamp – LED will blink while receiving MWI information.

CLIP

IncomingCall CLIP display – If you choose **No display**, no name and number of the incoming calls will be displayed. If you choose **Display number only**, just the number of the incoming calls will be displayed. If you choose **Display name and number**, then the name and number of all the incoming calls will be displayed on the phone set.



OutgoingCall CLIP hidden – If you choose **No hidden**, then the name and number of the outgoing calls will be displayed on remote end. If you choose **Hidden name only**, just the number of the outgoing call will be displayed. If you choose **Hidden name and number**, the name and number of all the outgoing calls will not be displayed on the phone set of remote end.



Call Park

Call Park Dial Number - It allows a person to put a call on hold at one telephone set and continue the conversation from any other telephone set. Such number you type here is determined by your ISP. The default Call Park Dial Number is “700”.

IncomingCall Rings

Rings as all ports in the group- All the ring ports selected for such account will ring when VigorTalk receives any incoming call.

Rings the first available port- The first ring port selected for such account will ring when VigorTalk receives any incoming call.

Rings by round robin- All the ring ports selected for such account will ring one by one when VigorTalk receives any incoming call.

Ring Port Setting

VigorTalk ATA-24 SH allows to connect up to 24 ring port. For such account, please specify required ring port(s) by checking the box(es) for applying the configuration.

3.5.2 Port Settings

Port Settings page allows users to set phone number and phone groups for different call receivers.

VoIP - Port Settings										
#	Edit	Active	SIP Account	Call Waiting	Hotline	Mic/Spk Gain	FAX	Codec	DTMF	Port Locked
1		V	1 - 1001			0 / 0	T.38 Relay	G.729A	RFC2833	
2		V	2 - 1002			0 / 0	T.38 Relay	G.729A	RFC2833	
3		V	3 - 1003			0 / 0	T.38 Relay	G.729A	RFC2833	
4		V	4 - 1004			0 / 0	T.38 Relay	G.729A	RFC2833	
5		V	5 - 1005			0 / 0	T.38 Relay	G.729A	RFC2833	
6		V	6 - 1006			0 / 0	T.38 Relay	G.729A	RFC2833	
7		V	7 - 1007			0 / 0	T.38 Relay	G.729A	RFC2833	
8		V	8 - 1008			0 / 0	T.38 Relay	G.729A	RFC2833	

- Edit** Click this button to access into the Edit page for each phone number.
- Active** Displays the status (active or not) for the VoIP connection. When this connection is active, a ‘v’ sign will be displayed on the page.
- SIP Account** Displays the account name for that port.
- Call Waiting** When call waiting is enabled, a ‘v’ sign will be displayed on the page.
- Hotline** Displays the hotline number for that port.
- Mic/Spk Gain** Displays the gain value for transmitting/receiving voice.
- FAX** Displays the FAX function mode, T.38 Relay or Transparent.
- Codec** Displays the codec settings for the VoIP connection.
- DTMF** Displays the DTMF mode (InBand, OutBand, SIP Info, etc.)
- Port Locked** When this port is locked, a ‘v’ sign will be displayed on the page. In general, it means the connection for such port is troubled with something.

When you click **Edit**, the following page will appear for you to configure.

VoIP - Port Settings - Port1 - Edit

Port 1

Disable Enable

Default SIP Accounts:

Hotline

Hotline Number to Internet:

Codec

Preferred Codec:

Single Codec:

Codec Rate: (ms)

Codec VAD: Disable Enable

CAS

Microphone Gain: (Range: -14 ~ 6)

Speaker Gain: (Range: -14 ~ 6)

FAX

FAX Mode:

DTMF

DTMF Mode: InBand OutBand(RFC2833) SIP INFO

DTMF Volume: (Range: 0 ~ 31)

Supplemental Service

Supplemental Service Mode: Disable Normal CHT

Supplemental Service Items: Call Waiting
 Call Transfer
 Call Conference

Do Not Disturb

Do Not Disturb: Disable Enable

Port Unlock

Manual Unlocked: (Unlock the port when its call status is "Line Fault")

Port 1

Click **Enable** to activate this port or **Disable** to close this port.
Default SIP Accounts – Choose one of the SIP account as the default setting.

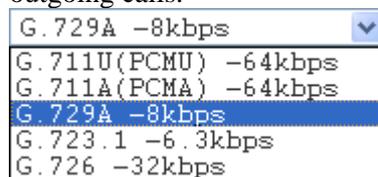
VoIP IP Address - The interface is used to apply VoIP traffics. There are two options: **WAN** and **LAN/VPN**. If LAN/VPN is selected, VoIP can be applied through a VPN tunnel to create a high security voice phone.

Hotline

Hotline Number to Internet - Pre-set a phone number to make the port dialing out to Internet automatically.

Codec

Preferred Codec - It can be applied on this port. VigorTalk ATA-24 SH supports five Codecs. The default setting is G.729A. You can choose another one as preferred Codec for outgoing calls.



Single Codec - If you checked this box, only preferred codec will be used for outgoing and incoming calls. And if the remote end does not support such Codec, the VoIP communication will be failed.

Codec Rate - Type the rate value to be applied on this port.

Codec VAD- Enable or Disable VAD (Voice Activity Detection). It can detect whether the voice activity is progressing or not. If not, RTP packets transmission will be stopped for saving more bandwidth.

CAS

Microphone Gain- The gain value while transmitting voice. The default value is 0. The range is from -32 to 31.

Speaker Gain- The gain value while receiving voice. The default value is 0. The range is from -32 to 31.

FAX

FAX Mode -The FAX function mode. There are three options:
Transparent: FAX will be transmitted via voice channel; no fax relay and no Codec change will be involved.

T.38 Relay: Using T.38 Fax Relay. This is the default value.

DTMF

DTMF Mode -

InBand: Choose this one then the Vigor will send the DTMF tone as audio directly when you press the keypad on the phone.

OutBand (RFC2833): Choose this one then the Vigor will capture the keypad number you pressed and transform it to digital form then send to the other side; the receiver will generate the tone according to the digital form it receive. This function is very useful when the network traffic congestion occurs and it still can remain the accuracy of DTMF tone.

SIP INFO: Choose this one then the Vigor will capture the DTMF tone and transfer it into SIP form. Then it will be sent to the remote end with SIP message.

DTMF Volume – Determine the volume of DTMF voice signal. The more the number is set, the greater the sound is.

Supplemental Service

Supplemental Service Mode – Additional services for VoIP in addition to the general communications.

Disable: The supplemental service will not be offered if choosing such item.

Normal: Means to enable the supplemental service with the general operation.

CHT: Means to enable the supplemental service with the operation based on the requirement from CHT.

Supplemental Service Items – There are three items - Call

Waiting, Call Transfer and Call Conference offered for you to check to activate the function(s).

Do Not Disturb

Click **Enable** to activate this function or **Disable** to close this function. If you click **Enable**, remote end cannot call such port.

Port Unlocked

This button is available only when current port is locked. Click it to unlock the port.

Apply

When you finish all the configurations, please click this button to activate them.

3.5.3 Speed Dial

This page allows you to set a simple way to dial a specific number. Up to 150 numbers can be stored in VigorTalk ATA-24 SH Series.

#	Speed Dial Phone Number	Speed Dial Destination	Memo
1	1001	1001@iptel.org	dial 1
2			
3			
4			
5			

Example 101 101@iptel.org

1 2 3 4 5 6 7 8 9 10 >

Apply Cancel Clear This Page

Speed Dial Phone Number Type the phone number to be used as quick dial.

Speed Dial Destination Type the destination address of the dial.

Memo Type a description for the specified number.

Apply Click this button to activate the page settings.

Clear This Page Click this button to remove all the settings in this page.

3.5.4 Dial Plan

Dial plan allows users to call out with simple buttons instead of dialing long numbers. To set a dial plan with specified settings, please open the following page.

VoIP - Dial Plan								
#	Match String	Min Length	Max Length	Prefix Strip	Prefix Add	SIP IP Address	Time Out	Memo
1								
2								
3								
4								
5								
6								
7								
8								
9								
10								

1

- Match String** Displays the match string of the entry.
- Min Length** Displays the min dial digit length of the entry.
- Max Length** Displays the max dial digit length of the entry.
- Prefix Strip** Displays the prefix string digit of the entry.
- Prefix Add** Displays the prefix add digit of the entry.
- SIP IP Address** Displays the SIP IP Address of the entry.
- Time Out** Displays the digit timeout value of the entry.
- Memo** Displays the brief description stated in memo field of the entry.
- Edit** Click this button to access into the editing page of the speed dial.
- Delete/Delete All** Click this button to delete the selected setting or all settings.

To configure one entry, please click **Edit** to open the following page.

VoIP - Dial Plan - Edit	
1	
Match String :	<input type="text"/>
Min Length:	<input type="text"/>
Max Length:	<input type="text"/>
Prefix Strip :	<input type="text"/>
Prefix Add :	<input type="text"/>
SIP IP Address :	<input type="text"/>
Inter Digit TimeOut:	<input type="text"/>
Memo :	<input type="text"/>

- Match String** Assign a match String for this entry. For example, suppose the match string is 12345. When dial 1234567, the digit will be matched of this entry.
- Min Length** Min length to match the string.
- Max Length** Max length to match this string
- Prefix Strip** Assign the length of digit to be removed from the original phone number. For example, suppose the original phone

number is 03654321 and the strip length is 2. The first two numbers (03) will be removed and the final phone number becomes 654321.

Prefix Add

Assign a new number to be added before the phone number (after removing length of digit). For example, suppose the original phone number is 03654321. The strip length is 2 and the append number is 886. Then, the final phone number will be 886654321..

SIP IP Address

Assign an IP address for the destination which the SIP message would be sent to.

Inter Digit Timeout

Assign a timeout value for the inter digit.

Memo

A description of this entry.

3.5.5 Tone Settings

It is provided for fitting the telecommunication custom for the local area of the adapter installed. Wrong tone settings might cause inconvenience for users. To set the sound pattern of the phone set, simply choose a proper region to let the system find out the preset tone settings and caller ID type automatically. Or you can adjust tone settings manually if you choose **User Defined**. TOn1, TOff1, TOn2 and TOff2 mean the cadence of the tone pattern. TOn1 and TOn2 represent sound-on; TOff1 and TOff2 represent the sound-off.

VoIP - Tone Settings

Region : Caller ID Type :

Tone Classification	Low Frequency(Hz)	High Frequency(Hz)	TOn1 (10msec)	TOff1 (10msec)	TOn2 (10msec)	TOff2 (10msec)
Dial tone	<input type="text" value="350"/>	<input type="text" value="440"/>	<input type="text" value="500"/>	<input type="text" value="0"/>	<input type="text" value="0"/>	<input type="text" value="0"/>
Ringing tone	<input type="text" value="440"/>	<input type="text" value="480"/>	<input type="text" value="0"/>	<input type="text" value="0"/>	<input type="text" value="200"/>	<input type="text" value="400"/>
Busy tone	<input type="text" value="480"/>	<input type="text" value="620"/>	<input type="text" value="0"/>	<input type="text" value="0"/>	<input type="text" value="50"/>	<input type="text" value="50"/>
Congestion tone	<input type="text" value="480"/>	<input type="text" value="620"/>	<input type="text" value="0"/>	<input type="text" value="0"/>	<input type="text" value="25"/>	<input type="text" value="25"/>

Setting CPT Tone Timer

Dial Tone : Busy Tone : Howler Tone : Ringing Tone :

Special Dial Tone : Call Waiting Tone : Congestion Tone : Reorder Tone :

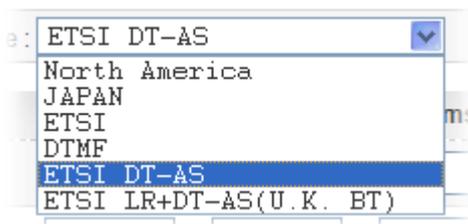
Region

Select the proper region which you are located. The common settings of **Caller ID Type**, **Dial tone**, **Ringing tone**, **Busy tone** and **Congestion tone** will be shown automatically on the page. If you cannot find out a suitable one, please choose **User Defined** and fill out the corresponding values for dial tone, ringing tone, busy tone, congestion tone by yourself for VoIP

phone.



Dial tone	A tone means the phone line is ready to make a call.
Ringing tone	A tone means the call is ringing.
Busy tone	A tone means the phone line is busy.
Congestion tone	A tone means the network is busy.
Low Frequency (Hz)	Type the low frequency number in Hertz.
High Frequency (Hz)	Type the high frequency number in Hertz.
TOn1 (10msec)	Type the duration of the first ring.
TOff1 (10msec)	Type the silence duration after the first ring.
TOn2 (10msec)	Type the duration of the next continuous ring.
TOff2 (10msec)	Type the silence duration after the next continuous ring.
Caller ID Type	If User Defined is selected in the Region field, users can select one of the supported values. If a country is selected, this field will display ID type value automatically.



Tone Timer	Set different timer for different tones to restrict the play time of tone. When the time is up, the tone broadcasting will be stopped. Dial Tone – A telephony signal which indicates that the status for the telephone is off-hook. Busy Tone – A telephony signal which indicates that the calling is failed.
-------------------	---

Howler Tone –A telephony signal to tell the caller the receiver is off-hook.

Ringling Tone - A telephony signal that the caller hears from the telephone set after dialing.

Special Dial Tone - A telephony signal which indicates a special feature (e.g., call forwarding) is using for such port.

Call Waiting Tone - A telephony signal which signifies that there is another incoming call, eg., call forwarding.

Congestion Tone – A telephony signal which indicates someone dials invalid call or the circuit (or network) is unable to route.

Recorder Tone – The caller has connected to an automatic answering device and is requested to start speaking.

3.5.6 Nat Traversal

If the router you use connects to Internet by other device, you have to set this function. This page is used to enable the Nat Traversal function. User could use it to enable the VoIP service under the NAT environment.

The screenshot shows the 'VoIP - NAT Traversal' configuration interface. It is divided into three main sections: 'NAT Traversal', 'Symmetric Media', and 'NAT Status'.
1. **NAT Traversal**: Contains three radio buttons: 'Disable' (selected), 'Manually Input NAT IP Address', and 'Auto Discover NAT IP Address'. Below these are input fields for 'NAT IP Address' (127.0.0.1), 'STUN Local Port' (3478), 'STUN Server Address' (stun.fwdnet.net), and 'STUN Server Port' (3478). There are also radio buttons for 'Semi-auto, need to config NAT' and 'Full-auto, no need to config NAT (only for SIP)'.
2. **Symmetric Media**: Contains two radio buttons: 'Disable symmetric RTP and T.38' (selected) and 'Enable symmetric RTP and T.38'.
3. **NAT Status**: Displays 'NAT Type: N/A, Local IP Address: 172.17.3.4, WAN IP Address: 172.17.3.4'.
At the bottom right, there are 'Apply' and 'Cancel' buttons.

Disable

The default setting is disable. If you want to enable this function, please click the other radio buttons listed below.

Manually Input NAT IP Address

To input NAT IP address manually, please click this radio button and type the IP Address in the NAT IP Address field.

Auto Discover NAT IP Address

To make the NAT IP address configuring automatically by the system, please click this radio button. There are two options for you to choose:

Semi-auto, need to config NAT – If you choose this one, you still have to configure NAT partially by type the required STUN Local Port, server address and server port.

Full-auto, no need to config NAT (only for SIP) – If you choose this one, the system will configure NAT settings automatically. All you have to do is click **Apply**.

Symmetric Media

If Symmetric Media is enabled, the router will transmit RTP packets/T.38 packets to the remote side via the IP address and Port number coming from the receiving RTP packet. If Symmetric Media is disabled, the router will transmit RTP packets via SDP (Session Description Protocol). The default setting is Disable Symmetric RTP and T.38.

NAT Status

Display current NAT status.

3.5.7 Line Test

This page is used to diagnose the connection status for device, port and subscriber line.



VoIP - Line Test

Port: 1 Line Test Function: Metallinc Loop Test Test Item: AC Voltage

Run Test

Port

Choose one of the VoIP port for executing line test.

Line Test Function

Choose one of the test functions for executing line test. There are three types provided here, loop, line card and user phone.

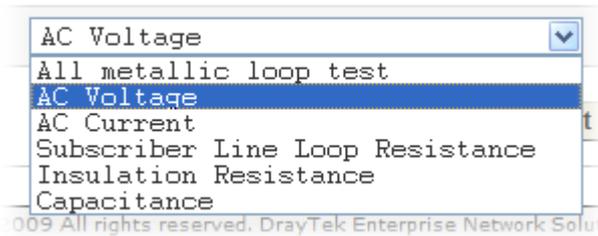
Metallinc Loop Test – Such function can diagnose if there is something wrong happened such as line cross, line short to ground and line open.

Line card Test – It can check normal battery, loop current and execute self dial tone test, self dial digit test and self ring voltage test.

User Phone Test – Such function must match with the subscriber.

Test Item

Choose one of the test types.



Run Test

Click this button to start the line test.

VoIP - Line Test Result

All Metallic Loop test (Port1)

AC Voltage
Tip Lead Voltage = 0 Vrms
Ring Lead Voltage = 0 Vrms
Tip Ring Voltage = 0 Vrms

AC Current
Tip Lead Current = 0.0 mA
Ring Lead Current = 0.0 mA

DC Voltage
Tip Lead Voltage = 0 V
Ring Lead Voltage = -0 V
Tip Ring Voltage = 0 V

DC Current
Tip Lead Current = 0 mA
Ring Lead Current = 0 mA

Subscriber Line Loop Resistance
RLOOP = 3809 Ohms

Resistance Test Results
RTG = OPEN
RRG = OPEN
RTR = 101040

Capacity
Tip Ground Capacitance = 0.00 uF
Ring Ground Capacitance = 0.00 uF
Tip Ring Capacitance = 0.42 uF

Receiver Off-Hook
Off-Hook = No

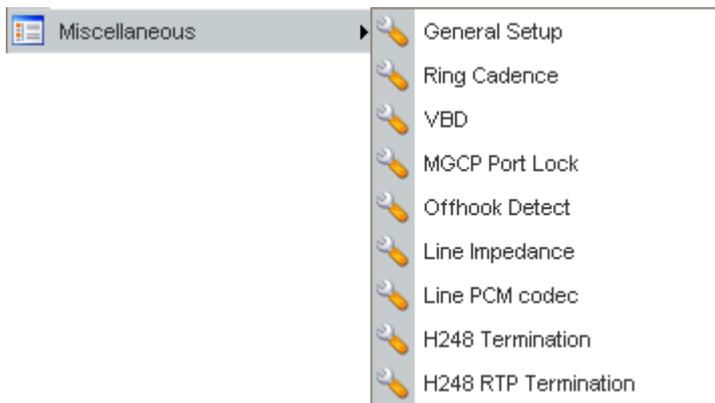
Ringer Equivalence Numer
REN = 0.465508
ZLoad = 15037.331055

Metallic Testing Result: Test Pass

[Return to line Test](#)

3.5.8 Miscellaneous

Many settings that cannot be classified under VoIP are placed in this page, such as ring cadence, voice band data, MGCP port lock, offhook detect, line impedance and line PCM codec.



General Setup

This page includes **RTP** and **T.38 Starting Port**, **T.38 Redundancy Number**, etc.

VoIP - Miscellaneous - General Setup

RTP Starting Port :	<input type="text" value="13456"/>
T38 Mode :	<input type="radio"/> Disable <input checked="" type="radio"/> Enable
T.38 Starting Port :	<input type="text" value="13456"/>
T.38 Redundancy number :	<input type="text" value="1"/> (Range: 0~4)
Dialing Completion Timeout :	<input type="text" value="4"/> sec (Range: 1~60)
Line Polarity Reversal :	<input type="checkbox"/> as Callee Answer <input type="checkbox"/> as Callee On-Hook
Echo Cancellation :	<input type="radio"/> Disable <input checked="" type="radio"/> Enable
Echo Cancellation Tail Length :	<input type="text" value="128"/> ms (Range: 8 ~ 128, should be multiple of 8)
Packet Loss Concealment :	<input checked="" type="radio"/> Disable <input type="radio"/> Enable

Jitter Buffer

Jitter Buffer Active	<input checked="" type="radio"/> Disable <input type="radio"/> Enable
Jitter buffer initial	<input type="text" value="0"/> ms (Range: 0 ~ 300ms)
Minimum jitter buffer size time	<input type="text" value="0"/> ms (Range: 0 ~ 300ms)
Maximum jitter buffer size time	<input type="text" value="200"/> ms (Range: 0 ~ 300ms)

Pulse Timing

Minimum pulse break time :	<input type="text" value="25"/> ms
Maximum pulse break time :	<input type="text" value="100"/> ms
Minimum flash break time :	<input type="text" value="250"/> ms
Maximum flash break time :	<input type="text" value="800"/> ms
Minimum pulse make time :	<input type="text" value="15"/> ms
Maximum pulse make time :	<input type="text" value="75"/> ms
Minimum pulse interdigit time :	<input type="text" value="250"/> ms

Required

1. Minimum pulse break time < Maximum pulse break time < Minimum flash break time < Maximum flash break time
2. Minimum pulse make time < Maximum pulse make time < Minimum pulse interdigit time

RTP Starting Port

The starting port number for RTP protocol packet. The default setting is 13456.

T38 Mode

Click **Enable** to enable T.38 function. Click **Disable** to close this function.

T.38 Starting Port

The starting port number for T.38 protocol packet. The default setting is 49170.

T.38 Redundancy Number

The redundancy number (how many payloads attaching to the tail of the packet) for T.38 protocol. The default value is 1.

Dialing Completion Timeout

Users might dial with incomplete phone number and wait for several seconds but not finish the complete dialing. The system will force to dial the incomplete number after the time you set in this field to finish that call. For example, the phone

number is 03654321 and the dialing completion timeout is set to 4 (secs). The user dials with 036 and stops to dial. After passing through 4 seconds, the adapter will send out that phone call automatically.

Line Polarity Reversal	as Callee Answer - Check this box to generate line polarity reversal while the remote user picks up the phone call. as Callee On-Hook - Check this box to generate line polarity reversal while the remote user hangs off the phone call.
Echo Cancellation	Click Enable to cancel echo. Click Disable to invoke echo.
Echo Cancellation Tail ...	The length is used to indicate the echo canceller buffer to cancel the echo. The unit is mini-second.
Packet Loss Concealment	It is a technique to mask the effects of packet loss in VoIP communications. Because the voice signal is sent as packets on a VoIP network, a packet might arrive very late, be corrupted or simply might not arrive to the receiver. Please enable this function. The device will try to cope with packet loss.
Jitter Buffer Active	Jitter Buffer is used to improve the quality of speech. It reorders the packets in the sequence they were sent and eliminates jitter at the cost of the overall delay. Click Enable to invoke such function. Click Disable to close such function.
Jitter buffer initial	It controls the initial delay of the jitter buffer (with unit of ms). The valid range is from 0 ms to 300 ms.
Minimum/Maximum jitter buffer size time	The Minimum/Maximum jitter values set the range for the Jitter buffer and control the minimum/maximum jitter buffer size (with unit of ms). The valid range is from 0 ms to 300 ms.
Minimum/Maximum pulse break time	Such setting determines the pulse break time for pulse digit phone. Keep the default setting.
Minimum/Maximum flash break time	Such setting determines the pulse break time for pulse digit phone. Keep the default setting.
Minimum/Maximum pulse make time	Such setting determines the pulse break time for pulse digit phone. Keep the default setting.
Minimum pulse interdigit time	Such setting determines the pulse break time for pulse digit phone. Keep the default setting.

Ring Cadence

This page is used to set ring cadence for each ring port. There are eight groups of ring cadence offered by the system.

VoIP - Miscellaneous - Ring Cadence									
#	Edit	Ton1	Toff1	Ton2	Toff2	Ton3	Toff3	Ton4	Toff4
1		1000 ms	4000 ms	0 ms	0 ms	0 ms	0 ms	0 ms	0 ms
2		800 ms	2000 ms	800 ms	4000 ms	0 ms	0 ms	0 ms	0 ms
3		800 ms	2000 ms	800 ms	4000 ms	0 ms	0 ms	0 ms	0 ms
4		800 ms	2000 ms	800 ms	4000 ms	0 ms	0 ms	0 ms	0 ms
5		800 ms	2000 ms	800 ms	4000 ms	0 ms	0 ms	0 ms	0 ms
6		800 ms	2000 ms	800 ms	4000 ms	0 ms	0 ms	0 ms	0 ms
7		800 ms	2000 ms	800 ms	4000 ms	0 ms	0 ms	0 ms	0 ms
8		800 ms	2000 ms	800 ms	4000 ms	0 ms	0 ms	0 ms	0 ms

Ton1 and Ton2 represent sound-on; Toff1 and Toff2 represent the sound-off.

To edit an entry, select it by clicking the radio button (from 1 to 8). Then click the **Edit** button on the bottom to bring up the following Web page.

VoIP - Ring Cadence - 1 - Edit	
Ton1 :	<input type="text" value="1000"/> ms
Toff1 :	<input type="text" value="4000"/> ms
Ton2 :	<input type="text" value="0"/> ms
Toff2 :	<input type="text" value="0"/> ms
Ton3 :	<input type="text" value="0"/> ms
Toff3 :	<input type="text" value="0"/> ms
Ton4 :	<input type="text" value="0"/> ms
Toff4 :	<input type="text" value="0"/> ms
<input type="button" value="Apply"/> <input type="button" value="Cancel"/>	

Ton1 Type the duration of the first ring.

Toff1 Type the silence duration after the first ring.

Ton2/Ton3/Ton4 Type the duration of the next continuous ring.

Toff2/Toff3/Toff4 Type the silence duration after the next continuous ring.

VBD

VBD means **Voice Band Data** which can determine Modem or Fax or Auto mode for data transmission according to the answering tone.

VoIP - Miscellaneous - Voice Band Data

All Ports
 One by One

Port 1:	Auto	Port 7:	Auto	Port 13:	Auto	Port 19:	Auto
Port 2:	Auto	Port 8:	Auto	Port 14:	Auto	Port 20:	Auto
Port 3:	Auto	Port 9:	Auto	Port 15:	Auto	Port 21:	Auto
Port 4:	Auto	Port 10:	Auto	Port 16:	Auto	Port 22:	Auto
Port 5:	Auto	Port 11:	Auto	Port 17:	Auto	Port 23:	Auto
Port 6:	Auto	Port 12:	Auto	Port 18:	Auto	Port 24:	Auto

All Port

When you choose Auto, Modem, Fax from the drop down menu, all the configurations will be applied to all ring ports.

Auto – Choose this setting to let the device determine which one (Modem or Fax) is proper.

Modem – Choose this setting to let the device sending the data through modem. When the device detects answer tone, it will force to use Modem mode.

Fax – Choose this setting to let the device sending the data by way of fax machine. When the device detects answer tone, it will force to use Fax mode.

One by One

When you click such item, you have to specify which ring port will be applied with the configuration set here. If you choose multiple ring ports, they will apply the configuration one by one.

MGCP Port Lock

Such device has the ability to detect error automatically. When something wrong happened, the system will lock all the MGCP ports. This page is available only when you choose **MGCP** as VoIP protocol.

VoIP - Miscellaneous - MGCP Port Lock

Port	Lock Status	Manual Control		Port	Lock Status	Manual Control	
1	Unlocked	<input type="button" value="Unlock"/>	<input type="button" value="Lock"/>	13	Unlocked	<input type="button" value="Unlock"/>	<input type="button" value="Lock"/>
2	Unlocked	<input type="button" value="Unlock"/>	<input type="button" value="Lock"/>	14	Unlocked	<input type="button" value="Unlock"/>	<input type="button" value="Lock"/>
3	Unlocked	<input type="button" value="Unlock"/>	<input type="button" value="Lock"/>	15	Unlocked	<input type="button" value="Unlock"/>	<input type="button" value="Lock"/>
4	Unlocked	<input type="button" value="Unlock"/>	<input type="button" value="Lock"/>	16	Unlocked	<input type="button" value="Unlock"/>	<input type="button" value="Lock"/>
5	Unlocked	<input type="button" value="Unlock"/>	<input type="button" value="Lock"/>	17	Unlocked	<input type="button" value="Unlock"/>	<input type="button" value="Lock"/>
6	Unlocked	<input type="button" value="Unlock"/>	<input type="button" value="Lock"/>	18	Unlocked	<input type="button" value="Unlock"/>	<input type="button" value="Lock"/>
7	Unlocked	<input type="button" value="Unlock"/>	<input type="button" value="Lock"/>	19	Unlocked	<input type="button" value="Unlock"/>	<input type="button" value="Lock"/>
8	Unlocked	<input type="button" value="Unlock"/>	<input type="button" value="Lock"/>	20	Unlocked	<input type="button" value="Unlock"/>	<input type="button" value="Lock"/>
9	Unlocked	<input type="button" value="Unlock"/>	<input type="button" value="Lock"/>	21	Unlocked	<input type="button" value="Unlock"/>	<input type="button" value="Lock"/>
10	Unlocked	<input type="button" value="Unlock"/>	<input type="button" value="Lock"/>	22	Unlocked	<input type="button" value="Unlock"/>	<input type="button" value="Lock"/>
11	Unlocked	<input type="button" value="Unlock"/>	<input type="button" value="Lock"/>	23	Unlocked	<input type="button" value="Unlock"/>	<input type="button" value="Lock"/>
12	Unlocked	<input type="button" value="Unlock"/>	<input type="button" value="Lock"/>	24	Unlocked	<input type="button" value="Unlock"/>	<input type="button" value="Lock"/>

Unlocked

When line error occurred, the system will lock all the troubled ports. It means all the locked ports will not be used any more. Users can execute line test to make sure if the troubled port is recovered to normal condition. If yes, users can open this web page to unlock those troubled ports.

Locked

Ports with locked status will not be used normally. In addition, any available MGCP port can be locked at any time due to special reason if necessary.

Offhook Detect

The value typed here can be used for the device to judge the time for offhook.

VoIP - Miscellaneous - Offhook Detect Current Value

All Ports 8 mA

One by One

Port 1: 8 mA	Port 7: 8 mA	Port 13: 8 mA	Port 19: 8 mA
Port 2: 10 mA	Port 8: 8 mA	Port 14: 8 mA	Port 20: 8 mA
Port 3: 8 mA	Port 9: 8 mA	Port 15: 8 mA	Port 21: 8 mA
Port 4: 8 mA	Port 10: 8 mA	Port 16: 8 mA	Port 22: 8 mA
Port 5: 8 mA	Port 11: 8 mA	Port 17: 8 mA	Port 23: 8 mA
Port 6: 8 mA	Port 12: 8 mA	Port 18: 8 mA	Port 24: 8 mA

Apply Cancel

All Port

When you click this button and choose any number from the drop down list, all the configurations will be applied to all ring ports.

8/10/12 – When the phone line current reaches 8mA/10mA/12mA, the system will judge the phone is off-hook.

One by One

When you click such item, you have to specify which ring port will be applied with the configuration set here. If you choose multiple ring ports, they will apply the configuration one by one.

Line Impedance

It defines the impedance of phone line for different areas (countries). At present, there are three types, 600, 900 and China (specified for areas in China) provided here for choosing.

VoIP - Miscellaneous - Line Impedance

All Ports 600 Ohm

One by One

Port 1: 600 Ohm	Port 7: 600 Ohm	Port 13: 600 Ohm	Port 19: 600 Ohm
Port 2: 600 Ohm	Port 8: 600 Ohm	Port 14: 600 Ohm	Port 20: 600 Ohm
Port 3: 600 Ohm	Port 9: 600 Ohm	Port 15: 600 Ohm	Port 21: 600 Ohm
Port 4: 600 Ohm	Port 10: 600 Ohm	Port 16: 600 Ohm	Port 22: 600 Ohm
Port 5: 600 Ohm	Port 11: 600 Ohm	Port 17: 600 Ohm	Port 23: 600 Ohm
Port 6: 600 Ohm	Port 12: 600 Ohm	Port 18: 600 Ohm	Port 24: 600 Ohm

Apply Cancel

All Port

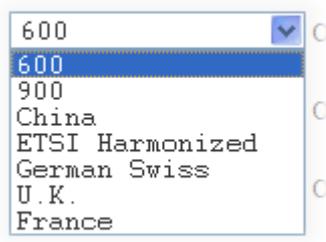
When you click this button and choose any item from the drop down list, all the configurations will be applied to all ports.

600/900 – Available impedance value provided by the system.

China, ETSI Harmonized, German Swiss, U.K, France -

Such selection is available for the users in China, ETSI

Harmonized, German Swiss, U.K or France.

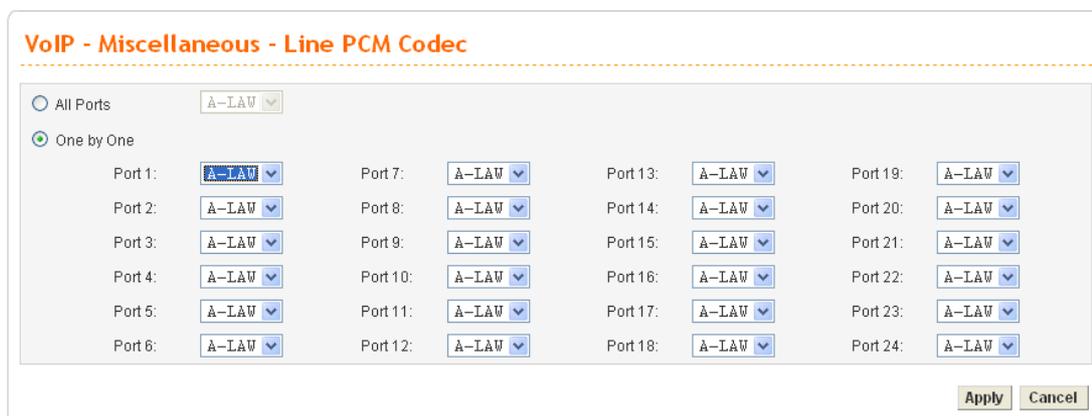


One by One

When you click such item, you have to specify which ring port will be applied with the configuration set here. If you choose multiple ports, they will apply the configuration one by one.

Line PCM codec

There are two types, **A-LAW** and μ **LAW** provided for such setting. Choose the suitable one according to the codec system used by ISP in different area. It will be applied for transferring analog signal into digital signal or transferring digital signal into analog signal while doing PCM codec sampling.



All Port

When you click this button and choose any item from the drop down list, all the configurations will be applied to all ports.

One by One

When you click such item, you have to specify which ring port will be applied with the configuration set here. If you choose multiple ports, they will apply the configuration one by one.

H248 Termination

Define the H248 Termination ID. There are 24 ports offered. Therefore, you have to define 24 Termination IDs. This value is available only H248 protocol is selected for VoIP.

VoIP - Miscellaneous - H248 Termination Setting

Termination ID Setting

All Termination ID Prefix ID Name : Start of ID Name Number :

One by One

1	<input type="text" value="AG1"/>	7	<input type="text" value="AG7"/>	13	<input type="text" value="AG13"/>	19	<input type="text" value="AG19"/>
2	<input type="text" value="AG2"/>	8	<input type="text" value="AG8"/>	14	<input type="text" value="AG14"/>	20	<input type="text" value="AG20"/>
3	<input type="text" value="AG3"/>	9	<input type="text" value="AG9"/>	15	<input type="text" value="AG15"/>	21	<input type="text" value="AG21"/>
4	<input type="text" value="AG4"/>	10	<input type="text" value="AG10"/>	16	<input type="text" value="AG16"/>	22	<input type="text" value="AG22"/>
5	<input type="text" value="AG5"/>	11	<input type="text" value="AG11"/>	17	<input type="text" value="AG17"/>	23	<input type="text" value="AG23"/>
6	<input type="text" value="AG6"/>	12	<input type="text" value="AG12"/>	18	<input type="text" value="AG18"/>	24	<input type="text" value="AG24"/>

All Termination ID

It allows you to configure the settings at one time with ID name and number.

Prefix ID Name – Type the name as termination ID. For example, if you type AG here, all the termination IDs (24 ports) will be set with the name of AG.

Start of ID Name Number – Type the number as a starting point. If you type 58900 here, the first termination ID will be given with the number of 58900 and the last termination ID will be given with the number of 58923.

One by One

Type the termination ID settings manually and respectively.

H248 RTP Termination

The RTP termination is a special termination that is created when a call is started. It will be removed once the call is completed. The RTP termination ID has the format "<Prefix><digit>". If <prefix> is "RTP", and <digit> is "1000", then RTP termination ID will be "RTP1000". H248 Client will reply the RTP termination ID to the call agent while it tries to create a call. The <digit> part of RTP termination ID is increased with one for each call.

VoIP - Miscellaneous - H248 RTP Termination ID

Start Number :

Count Number :

Digit Length :

Prefix :

Start Number

Set the start number of digit string.

- Count Number** Set the number of RTP termination ID.
- Digit Length** Set the length of the digit string.
- Prefix** Set the prefix name for the RTP termination ID.

For example:

Set <Start Number>=0, <Count Number>=24, <Digit Length>=5, <Prefix>="RTP".
 The RTP termination ID will be **RTP/00000, RTP/00001, RTP/00002, RTP/00023.**

Set <Start Number>=1, <Count Number>=100, <Digit Length>=2, <Prefix>="RTP".
 The RTP termination ID will be **RTP01, RTP02, RTP03, RTP100.**

3.5.9 Incoming Call Barring

This feature is used to bar incoming VoIP calls from the Internet. Barring classes can be specified to allow or deny incoming calls. There are five barring classes on the device. The default setting is **Allow all incoming calls**.



Set

This page allows you to choose a barring class, match method and set a range for speed dial entries for the incoming call barring.

VoIP - Incoming Call Barring - Set

Barring Class

Deny only calls from deny list ▼

Match Method

Name: Disable Enable
 Remind:

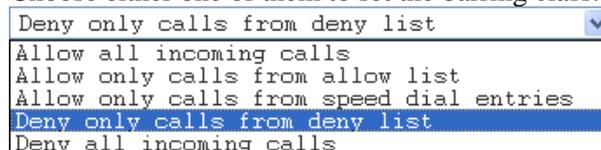
IP/Domain: Disable Enable
 Remind:

Speed Dial Entries

From: 1 ▼ To: 150 ▼

Barring Class

There are five options for incoming calls from remote ends. Choose either one of them to set the barring class.



Allow all incoming calls – All incoming calls from remote ends are accepted by this adapter.

Allow only calls from allow list – Only the calls listed in the Allow List page will be accepted by this adapter.

Allow only calls from speed dial entries – Only the calls listed in the speed dial entries will be accepted by this adapter.

Deny only calls from deny list – The calls listed on Deny List page will not be accepted by this adapter. And others calls are accepted.

Deny all incoming calls – All incoming calls from remote ends are not accepted by this adapter.

Match Method

Name - Enable or Disable this function to take value of **Speed Dial Phone Number** to be checked.

IP/Domain - Enable or Disable this function to take the value of **Speed Dial Destination** to be checked.

Speed Dial Entries

Type the range to be checked. The default value is from 1 to 150.

Allow List

The VigorTalk ATA-24 SH supports up to **30** entries in the Allow List table. When you choose **Allow only calls from allow list** as the Barring Class, only the people listed in this list can call this adapter.

#	Name	IP/Domain
1	<input type="text" value="Tom"/>	<input type="text" value="192.168.1.6"/>
2	<input type="text" value="John"/>	<input type="text" value="iptel.org"/>
3	<input type="text"/>	<input type="text"/>
4	<input type="text"/>	<input type="text"/>
5	<input type="text"/>	<input type="text"/>
Example	John	192.168.1.1 or iptel.org

123456

Name

The name or number in the allow list.

IP/Domain

The IP address or domain name to be allowed. If the peer is registered in SIP proxy server, use the domain name of the SIP proxy server. Otherwise, use the static IP address or DDNS domain name.

Deny List

The VigorTalk ATA-24 SH supports up to **30** entries in the Deny List table. When you choose **Deny only calls from deny list** as the Barring Class, people listed in this list **cannot** call this adapter.

VoIP - Incoming Call Barring - Deny List

#	Name	IP/Domain
1	<input type="text" value="James"/>	<input type="text" value="172.16.3.221"/>
2	<input type="text" value="Steven"/>	<input type="text" value="arctel.com"/>
3	<input type="text"/>	<input type="text"/>
4	<input type="text"/>	<input type="text"/>
5	<input type="text"/>	<input type="text"/>
Example	John	192.168.1.1 or iptel.org

1 2 3 4 5 6

Name

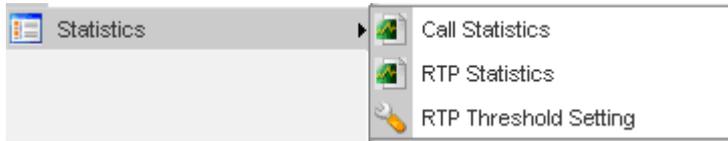
The name or number in the deny list.

IP/Domain

The IP address or domain name to be denied. If the peer is registered in SIP proxy server, use the domain name of the SIP proxy server. Otherwise, use the static IP address or DDNS domain name.

3.5.10 Statistics

The function provides call statistics, RTP statistics, and RTP threshold setting for users.



Call Statistics

This page displays statistics for all incoming/outgoing calls (successful and failed) through this adapter.

VoIP - Call Statistics

Refresh Option:

15 minutes | 24 hours

Port	Successful Outgoing Calls	Successful Incoming Calls	Failed Outgoing Calls	Failed Incoming Calls	Total Calls
1					
2					
3					
4					
5					
6					
7					
8					
9					
10					
11					
12					
13					
14					
15					
16					
17					
18					
19					
20					
21					
22					
23					
24					

You can click **Refresh** to get the latest status information for these VoIP phones. In addition, you can set the time interval of refreshing. Use the drop down list of **Refresh Option** to choose an automatic refreshing setting. If you choose **No Refresh**, the system will not refresh this page until you click **Refresh** button.



RTP Statistics

This page displays statistics for RTP.

VoIP - RTP Statistics

Refresh Option:

15 minutes | 24 hours

Port	Sender Packet Count	Sender Octet Count	Receiver Packet Count	Receiver Octec Count	Number of Packet Lost	Interarrival Jitter	Delay
1							
2							
3							
4							
5							
6							
7							
8							
9							
10							
11							
12							
13							
14							
15							
16							
17							
18							
19							
20							
21							
22							
23							
24							

You can click **Refresh** to get the latest status information for these VoIP phones. In addition, you can set the time interval of refreshing. Use the drop down list of **Refresh Option** to choose an automatic refreshing setting. If you choose **No Refresh**, the system will not refresh this page until you click **Refresh** button.



RTP Threshold Setting

This page is used to set RTP threshold settings for alert message. The alert message will be sent out when the values of the incoming phone calls beyond the settings configured in this page. In addition, the alert message will be displayed on the page of **Show Alert**.

VoIP - Statistics - RTP threshold

Mode : Disable Enable

Round Trip Delay Low Threshold : (ms)

Round Trip Delay High Threshold : (ms)

Jitter Low Threshold : (ms)

Jitter High Threshold : (ms)

Packet Loss Ratio Low Threshold : (0~100%)

Packet Loss Ratio High Threshold : (0~100%)

RTCP Timeout : (seconds)

- Mode** Click **Enable** to activate RTP Threshold mode.
- Round Trip Delay Low Threshold** Set the lowest value (default setting is 80) as round trip delay low threshold.
- Round Trip Delay High Threshold** Set the highest value (default setting is 150) as round trip delay high threshold.
- Jitter Low Threshold** Set the lowest value (default setting is 3) as jitter low threshold.
- Jitter High Threshold** Set the lowest value (default setting is 10) as jitter high threshold.
- Packet Loss Ratio Low Threshold** Set the lowest value (default setting is 0) as packet loss ratio low threshold.
- Packet Loss Ratio High Threshold** Set the lowest value (default setting is 5) as packet loss ratio high threshold.
- RTCP Timeout** Set the value (default setting is 10) for RTP timeout setting.

3.5.11 Status

This feature displays Port and SIP status for VoIP phone calls.



Port Status

This page displays the connection status for VoIP phone calls.

VoIP -Port Status

Refresh Option:

#	Call Status	Call Type	Caller Number	Callee Number	Start Time	Remote RTP Address	Remote RTP Port	RTP Statistic	Codec Type	Packet Period	VAD	DTMF Relay
1	Idle											
2	Idle											
3	Idle											
4	Idle											
5	Idle											
6	Idle											
7	Idle											
8	Idle											
9	Idle											
10	Idle											
11	Idle											
12	Idle											
13	Idle											
14	Idle											
15	Idle											
16	Idle											
17	Idle											
18	Idle											
19	Idle											
20	Idle											
21	Idle											
22	Idle											
23	Idle											
24	Idle											

* PS: Packets Sent, OS: Octets Sent, PR: Packets Received, OR: Octets Received, PL: Packets Lost, JI: Interarrival Jitter Estimate(ms), LA: Avg TX Delay(ms)

- Call Status** Display the calling status, idle, far-end alerting, alerting, busy, dialing and connected.
- Call Type** The dialing direction for this call (Incoming/Outgoing).
- Caller Number** The phone number of the caller.
- Callee Number** The phone number of the receiver.
- Start Time** The starting time of the call.
- Remote RTP Address** The IP address of remote voice site.
- Remote RTP Port** The used port number of remote voice site.
- RTP Statistic** The statistics for RTP. PS means packets sent; OS means octets sent; PR means packet received; OR means octets received; PL means packets lost, LA means average TX delay (unit is ms) and JI means inter arrival jitter estimates (unit is ms).
- Codec Type** The Codec mode used for this phone calling.
- Packet Period** The period of time for sampling on voice signal.
- VAD** The status of VAD.
- DTMF Relay** The status of DTMF.

You can click **Refresh** to get the latest status information for these VoIP phones. In addition, you can set the time interval of refreshing. Use the drop down list of **Refresh Option** to choose an automatic refreshing setting. If you choose **No Refresh**, the system will not refresh this page until you click **Refresh** button.



SIP Status

This page displays the registration status for VoIP phone calls.

VoIP -SIP Status

Refresh Option:

#	Register Status	#	Register Status	#	Register Status	#	Register Status
1		9		17		25	
2		10		18		26	
3		11		19		27	
4		12		20		28	
5		13		21		29	
6		14		22		30	
7		15		23		31	
8		16		24		32	

Register Status

The status (OK or Failed) of registering in proxy server.

You can click **Refresh** to get the latest status information for these VoIP phones. In addition, you can set the time interval of refreshing. Use the drop down list of **Refresh Option** to choose an automatic refreshing setting. If you choose **No Refresh**, the system will not refresh this page until you click **Refresh** button.



Fault Status

This page displays the status for VoIP fault. When something wrong happened to the VoIP line, the problem will be displayed in this page.

VoIP - Fault Status

Refresh Option:

Device

No Fault

Port	Fault Status						
1	-	7	-	13	-	19	-
2	-	8	-	14	-	20	-
3	-	9	-	15	-	21	-
4	-	10	-	16	-	22	-
5	-	11	-	17	-	23	-
6	-	12	-	18	-	24	-

Fault Status

The possible messages for the fault status include:

Thermal Fault: When the SLIC is too hot to be born, corresponding message will be displayed in this field.

DC Fault: DC current is added on the telephone line externally.

AC Fault: AC current is added on the telephone line externally.

Buttery Fault: there is something wrong happened to the internal battery.

Clock Fault: there is something wrong happened to the internal clock.

You can click **Refresh** to get the latest status information for these VoIP phones. In addition, you can set the time interval of refreshing. Use the drop down list of **Refresh Option** to choose an automatic refreshing setting. If you choose **No Refresh**, the system will not refresh this page until you click **Refresh** button.

No Refresh

No Refresh

Every 10 Seconds

Every 20 Seconds

Every 30 Seconds

3.5.12 Call History

This page lists the call history through VigorTalk ATA-24. You can click **Refresh** to get the latest history information for these VoIP phones. Besides, this page refreshes automatically every 10 seconds.

VoIP - Call History																
Refresh Option: No Refresh Refresh																
#	Port Number	Call Type	Caller Number	Callee Number	Start Time	End Time	Duration	Release Reason	Remote RTP Address	Remote RTP Port	RTP Statistic	Codec Type	Packet Period	VAD	DTMF Relay	
1	11	Incoming	2435	1011	Fri Jan 2 03:40:54 1970	Fri Jan 2 03:41:23 1970	0 days, 00h:00m:29s	Normal Drop	10.10.10.60	13476	PS=751, OS=15020, PR=605, OR=12120, PL=0, JI=0, LA=2	G.729A	8kbps	20ms	Off	RFC2833
2	20	Incoming	2444	1020	Fri Jan 2 03:40:54 1970	Fri Jan 2 03:41:23 1970	0 days, 00h:00m:29s	Normal Drop	10.10.10.60	13494	PS=751, OS=15020, PR=605, OR=12120, PL=0, JI=0, LA=2	G.729A	8kbps	20ms	Off	RFC2833
3	2	Incoming	2426	1002	Fri Jan 2 03:40:54 1970	Fri Jan 2 03:41:23 1970	0 days, 00h:00m:29s	Normal Drop	10.10.10.60	13458	PS=751, OS=15020, PR=605, OR=12100, PL=0, JI=0, LA=2	G.729A	8kbps	20ms	Off	RFC2833
4	19	Incoming	2443	1019	Fri Jan 2 03:40:49 1970	Fri Jan 2 03:41:18 1970	0 days, 00h:00m:29s	Normal Drop	10.10.10.60	13492	PS=1001, OS=20020, PR=885, OR=17360, PL=0, JI=0, LA=3	G.729A	8kbps	20ms	Off	RFC2833

Refresh Option

Specify the interval of refresh time to obtain the latest VoIP calling information. The information will update immediately when the Refresh button is clicked.



Port Number

The port number of VoIP.

Call Type

The dialing direction for this call (Incoming/Outgoing).

Caller Number

The phone number of the caller.

Callee Number

The phone number of the receiver.

Start Time

The starting time of the call.

End Time

The ending time of the call.

Duration

The duration of the call.

Release Reason

The reason for the call termination.

Remote RTP Address

The IP address of remote voice site.

Remote RTP Port

The used port number of remote voice site.

RTP Statistic

The statistic of RTP with abbreviation will be shown in this field (e.g., PS: Packets Sent; OS: Octets Sent; PR: Packets Received; OR: Octets Received; PL: Packets Lost; JI: Interarrival Jitter Estimate (ms); LA: Average TX Delay(ms)).

Codec Type

The Codec mode used for this phone calling.

Packet Period	The period of time for sampling on voice signal.
VAD	The status of VAD.
DTMF Relay	The status of DTMF.

3.5.13 Configure Activate

This page will activate the new configured settings. Click **Apply** to execute the new settings.

VoIP - Configure Activate

Warning !
The action may cause **all of the VoIP calls disconnected !**
Please confirm you really want to execute Configure Activate right now !

When the VoIP settings are configured, it must be activated after clicking **Apply** in this page.

4

Trouble Shooting

This section will guide you to solve abnormal situations if you cannot access into the Internet after installing the adapter and finishing the web configuration. Please follow below sections to check your basic installation stage by stage.

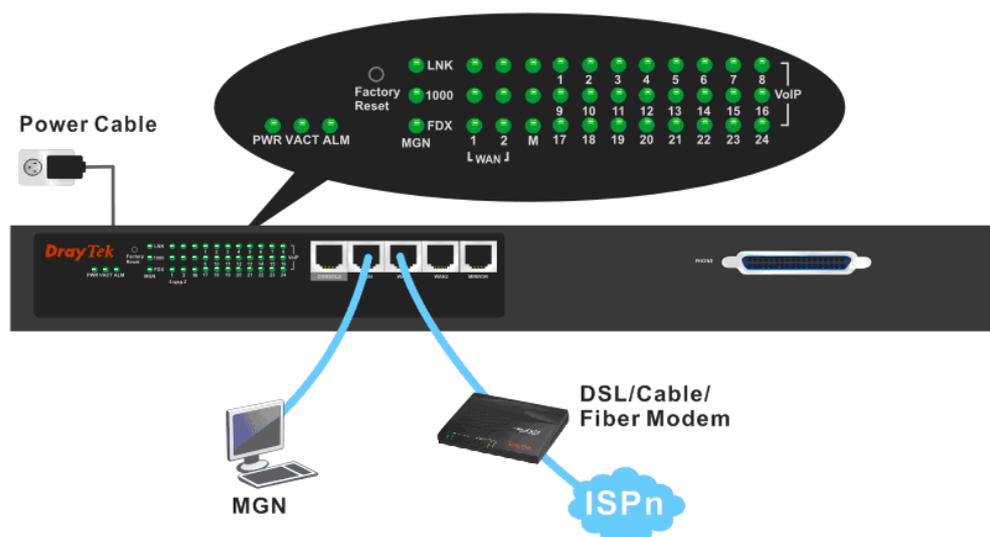
- Checking if the hardware status is OK or not.
- Checking if the Network Connection Settings on your computer is OK or not.
- Pinging the Adapter from your computer.
- Checking if the ISP Settings are OK or not.
- Backing to factory default setting if necessary.

If all above stages are done and the adapter still cannot run normally, it is the time for you to contact with your dealer for advanced help.

4.1 Checking If the Hardware Status Is OK or Not

Follow the steps below to verify the hardware status.

1. Check if the power line and WLAN/MGN cable connections is OK. If not, refer to “**2.1 Hardware Installation**” for reconnection.
2. Turn on the adapter. Make sure the **ACT LED** blinks once per second and the correspondent **WAN/MGN LED** is bright.



3. If not, there must be something wrong with the hardware connection. Simply back to “**1.2 Hardware Installation**” to execute the hardware installation. And then, try again.

4.2 Checking If the Network Connection Settings on Your Computer Is OK or Not

Sometimes the link failure occurs due to the wrong network connection settings. After trying the above section, if the link is still failed, please do the steps listed below to make sure the network connection settings is OK.

For Windows

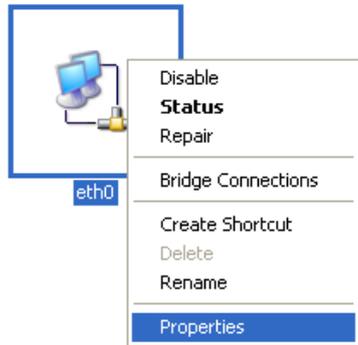


The example is based on Windows XP. As to the examples for other operation systems, please refer to the similar steps or find support notes in www.draytek.com.

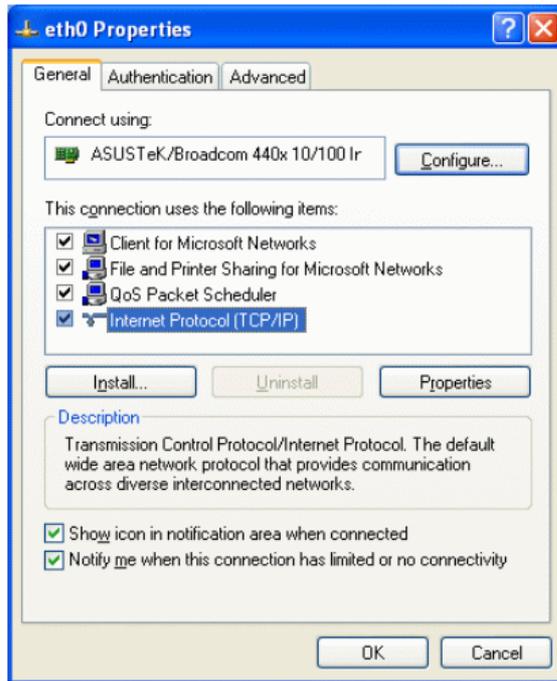
1. Go to Control Panel and then double-click on Network Connections.



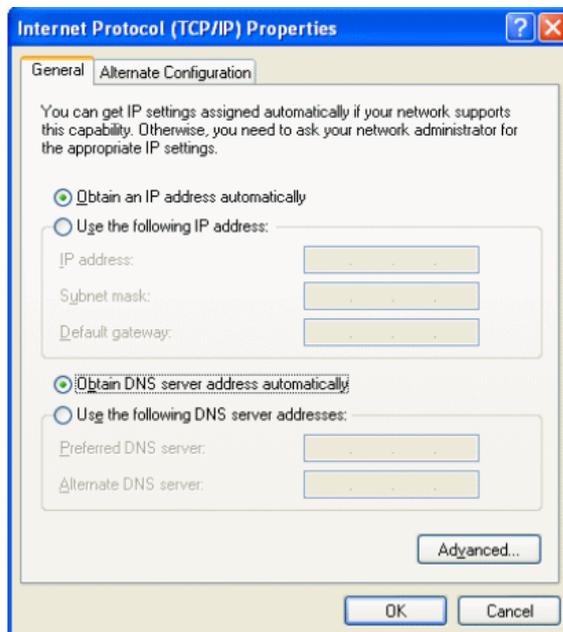
2. Right-click on Local Area Connection and click on Properties.



3. Select **Internet Protocol (TCP/IP)** and then click **Properties**.

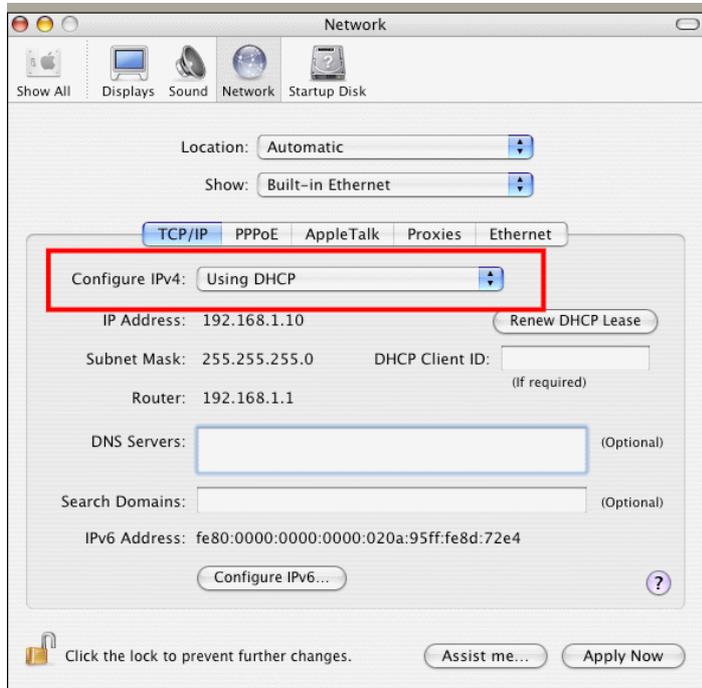


4. Select **Obtain an IP address automatically** and **Obtain DNS server address automatically**.



For MacOs

1. Double click on the current used MacOs on the desktop.
2. Open the **Application** folder and get into **Network**.
3. On the **Network** screen, select **Using DHCP** from the drop down list of Configure IPv4.



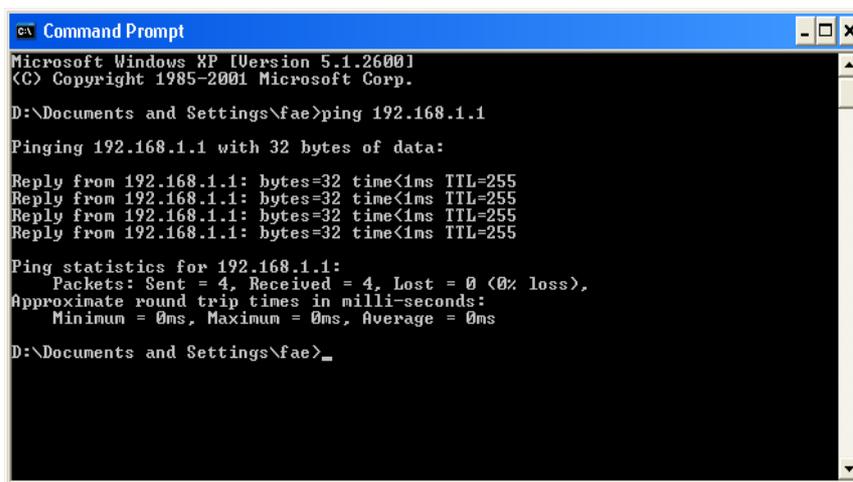
4.3 Pinging the Adapter from Your Computer

The default gateway IP address of the adapter is 192.168.1.1. For some reason, you might need to use “ping” command to check the link status of the adapter. **The most important thing for this command is that the computer will receive a reply from 192.168.1.1 for correct link.** If not, please check the IP address of your computer. We suggest you setting the network connection as **get IP automatically**. (Please refer to the section 3.2)

Please follow the steps below to ping the adapter correctly.

For Windows

1. Open the **Command Prompt** window (from **Start menu>> Run**).
2. Type **command** (for Windows 95/98/ME) or **cmd** (for Windows NT/ 2000/XP/Vista). The DOS command dialog will appear.



```
ca Command Prompt
Microsoft Windows XP [Version 5.1.2600]
(C) Copyright 1985-2001 Microsoft Corp.

D:\Documents and Settings\fae>ping 192.168.1.1

Pinging 192.168.1.1 with 32 bytes of data:

Reply from 192.168.1.1: bytes=32 time<1ms TTL=255

Ping statistics for 192.168.1.1:
    Packets: Sent = 4, Received = 4, Lost = 0 (0% loss),
    Approximate round trip times in milli-seconds:
        Minimum = 0ms, Maximum = 0ms, Average = 0ms

D:\Documents and Settings\fae>_
```

3. Type **ping 192.168.1.1** and press [Enter]. If the link is OK, the line of **Reply from 192.168.1.1:bytes=32 time<1ms TTL=255** will appear.
4. If the line does not appear, please check the IP address setting of your computer.

For MacOs (Terminal)

1. Double click on the current used MacOs on the desktop.
2. Open the **Application** folder and get into **Utilities**.
3. Double click **Terminal**. The Terminal window will appear.
4. Type **ping 192.168.1.1** and press [Enter]. If the link is OK, the line of **64 bytes from 192.168.1.1: icmp_seq=0 ttl=255 time=xxxx ms** will appear.

```

Terminal — bash — 80x24
Last login: Sat Jan 3 02:24:18 on ttty1
Welcome to Darwin!
Vigor10:~ draytek$ ping 192.168.1.1
PING 192.168.1.1 (192.168.1.1): 56 data bytes
64 bytes from 192.168.1.1: icmp_seq=0 ttl=255 time=0.755 ms
64 bytes from 192.168.1.1: icmp_seq=1 ttl=255 time=0.697 ms
64 bytes from 192.168.1.1: icmp_seq=2 ttl=255 time=0.716 ms
64 bytes from 192.168.1.1: icmp_seq=3 ttl=255 time=0.731 ms
64 bytes from 192.168.1.1: icmp_seq=4 ttl=255 time=0.72 ms
^C
--- 192.168.1.1 ping statistics ---
5 packets transmitted, 5 packets received, 0% packet loss
round-trip min/avg/max = 0.697/0.723/0.755 ms
Vigor10:~ draytek$

```

4.4 Checking If the ISP Settings Are OK or Not

1. Go to the web configuration GUI (<http://192.168.1.1>), click **Network >> WAN** to check your ISP settings for IP modes.
2. Make sure the **Active** check box has been selected.

Backup :	<input checked="" type="radio"/> Disable	<input type="radio"/> Enable		
#	Edit	IP Mode	Active	Defat
WAN1		PPPoE	<input checked="" type="checkbox"/>	<input checked="" type="radio"/>
WAN2		Not Set	<input type="checkbox"/>	<input type="radio"/>

For Static Mode

1. Check if the values of **IP Address**, **Subnet Mask**, **Gateway IP Address** and **Primary DNS** that you got from ISP are set properly or not. If you forget, please contact with ISP for getting new ones.

Static/DHCP Configuration

<table style="width: 100%; border-collapse: collapse;"> <tr> <td style="padding: 5px;">IP Address :</td> <td style="border: 1px solid #ccc; padding: 5px;">172.16.3.229</td> </tr> <tr> <td style="padding: 5px;">Subnet Mask :</td> <td style="border: 1px solid #ccc; padding: 5px;">255.255.255.0</td> </tr> <tr> <td style="padding: 5px;">Default Gateway :</td> <td style="border: 1px solid #ccc; padding: 5px;">172.16.3.1</td> </tr> <tr> <td style="padding: 5px;">Primary DNS :</td> <td style="border: 1px solid #ccc; padding: 5px;">168.95.1.1</td> </tr> <tr> <td style="padding: 5px;">Secondary DNS :</td> <td style="border: 1px solid #ccc; padding: 5px;">168.95.192.1</td> </tr> </table>	IP Address :	172.16.3.229	Subnet Mask :	255.255.255.0	Default Gateway :	172.16.3.1	Primary DNS :	168.95.1.1	Secondary DNS :	168.95.192.1	<p>Host Name : <input style="width: 100%;" type="text"/></p> <p>Domain Name : <input style="width: 100%;" type="text"/></p> <p><small>(Host Name and Domain Name are required for some ISPs.)</small></p>
IP Address :	172.16.3.229										
Subnet Mask :	255.255.255.0										
Default Gateway :	172.16.3.1										
Primary DNS :	168.95.1.1										
Secondary DNS :	168.95.192.1										

2. If anything wrong, please retype correct values and try the network connection again.
3. After finishing the settings, go to **System - Status** page and click **WAN Status**. You will get a correct web page of WAN settings.

WAN1 :	
IP Address :	220.130.52.221
MAC Address :	00:50:7f:28:80:e4
Primary DNS :	168.95.1.1
Secondary DNS :	
Gateway :	220.130.52.209
RX Packets :	708
TX Packets :	384
Connection Status :	connected
Up Time :	0 days 0 hours 5 minutes 7 seconds

For DHCP Mode

1. Check if **Host Name** (optional) and **Domain Name** (optional) are correct or not. Both them are required for some ISPs.

Static/DHCP Configuration			
IP Address :	<input type="text" value="172.16.3.229"/>	Host Name :	<input type="text"/>
Subnet Mask :	<input type="text" value="255.255.255.0"/>	Domain Name :	<input type="text"/>
Default Gateway :	<input type="text" value="172.16.3.1"/>	(Host Name and Domain Name are required for some ISPs.)	
Primary DNS :	<input type="text"/>		
Secondary DNS :	<input type="text"/>		

2. If anything wrong, please check and retype correct values. Then try the network connection again.
3. After finishing the settings, go to **System - Status** page and click **WAN Status**. You will get a correct web page of WAN settings.

WAN1 :	
IP Address :	172.16.100.10
MAC Address :	00:50:7f:28:80:e5
Primary DNS :	172.16.100.1
Secondary DNS :	
Gateway :	172.16.100.1
RX Packets :	96
TX Packets :	100
Connection Status :	connected
Up Time :	0 days 0 hours 4 minutes 51 seconds

4.5 Backing to Factory Default Setting If Necessary

Sometimes, a wrong connection can be improved by returning to the default settings. Try to reset the adapter by software or hardware.



Warning: After pressing **factory default setting**, you will lose all settings you did before. Make sure you have recorded all useful settings before you pressing. The password of the factory default is null.

Software Reset

You can reset adapter to factory default via Web page.

Go to **System >> Reboot** on the web page. The following screen will appear. Choose **Reset to factory default** and click **Apply**. After few seconds, the adapter will return all the settings to the factory settings.

System - Reboot

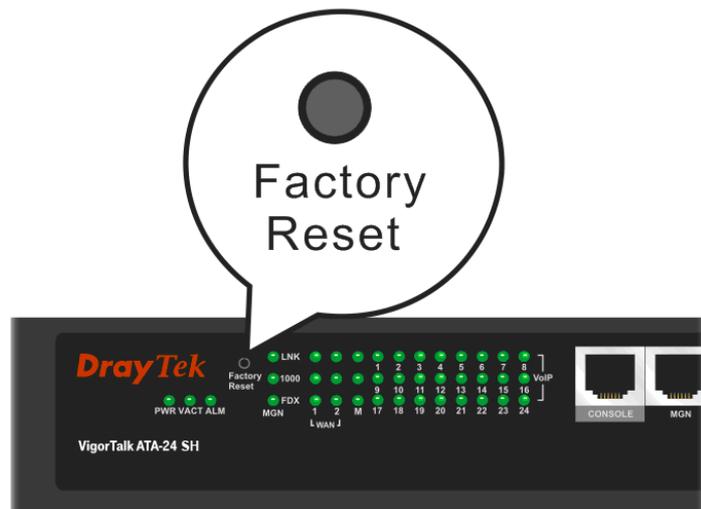
System rebooting will take 60 seconds

Reset to factory default

Apply

Hardware Reset

While the adapter is running (VACT LED blinking), press the **Factory Reset** button and hold for more than 5 seconds. Then, the adapter will restart with the default configuration.



After restore the factory default setting, you can configure the settings for the adapter again to fit your personal request.

4.6 Contacting Your Dealer

If the adapter settings are correct at all, and the adapter still does not connect to internet, please contact your ISP technical support representative to help you for configuration.

Also, if the adapter still cannot work correctly, please contact your dealer for help. For any further questions, please send e-mail to [**support@draytek.com**](mailto:support@draytek.com).

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Appendix A: Telnet Commands

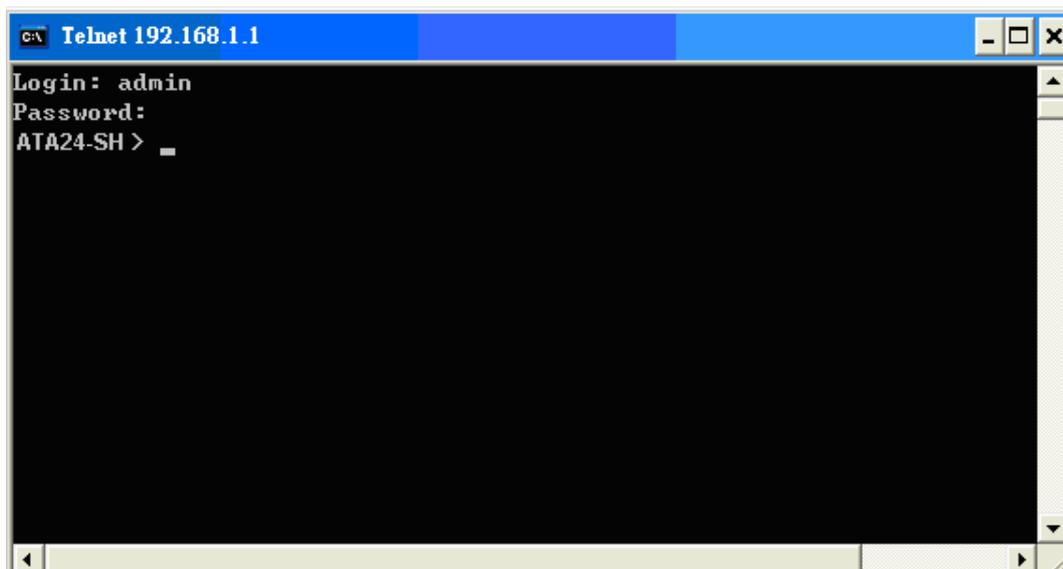
A.1 Introduction

In addition to the SNMP management, users can use commands to configure the ATA-24 VoIP Board. Users can do telnet on the ATA-24 VoIP Board and use the following two ways. One is console interface; another is telnet by management port.

The ATA-24 console interface will connect to PC console port. Users can use terminal emulation software configured by the following parameters.

- VT100 terminal emulation
- 115200 bps
- No parity, 8 data bits, 1 stop bit
- No hardware flow control

Users can type '?' for help. Another tools for command interface is telnet via management port. The PC should be the same subnet as ATA-24 VoIP Board. The default IP address is **192.168.1.1**. The default login name is "**admin**", password is "**1234**".



A.2 Root Commands

A.2.1 Enter Function Commands

- Enter advanced configuration function
ATA24-SH> advance
- Enter system diagnostics function
ATA24-SH> diag
- Enter firewall configuration function
ATA24-SH> firewall

- Enter network configuration function
ATA24-SH> network
- Enter system configuration function
ATA24-SH> system
- Enter voip configuration function
ATA24-SH> voip

A.2.2 Other Commands

- Help
ATA24-SH> ?
- Logout the CLI or the Telnet connection
ATA24-SH> exit
or
ATA24-SH> logout
or
ATA24-SH> quit

A.3 Advance Commands

A.3.1 General Commands

- Enter advance configuration function
ATA24-SH> advance
- Help in advance configuration function
ATA24-SH/ advance > ?
- Back to the root commands
ATA24-SH/ advance > ..

A.3.2 Port Block Commands

- Display the status for port block setting
ATA24-SH/advance> block -s
- Enable port block setting
ATA24-SH/advance> <Index> <Enable> <Port number>
- Disable port block setting
ATA24-SH/advance> <Index> <Disable>

<Index>	Item number(1~10)
<Disable/Enable>	0: Disable 1: Enable
<Port Number>	Available number 1 ~ 65535

A.3.3 Portmirror Commands

- Help
ATA24-SH/advance> portmirror ?
- Display port mirror settings
ATA24-SH/advance> portmirror -s
- Edit port mirror settings
ATA24-SH/advance> <Enable> <Moirroring> <Mirror CPU> <Mirror LAN><Mirror WAN1> <Mirror WAN2> <Mirror WAN3>

<Enable>	0: Disable 1: Enable
<Moirroring>	Moirroring Port 1: WAN1 2: WAN2
<Mirror CPU> <Mirror LAN> <Mirror WAN1> <Mirror WAN2> <Mirror WAN3>	0: Do not mirror, 1: Mirror

A.4 Diagnostics Commands

A.4.1 General Commands

- Enter system diagnostics function
ATA24-SH> diag
- Help in the system diagnostics function
ATA24-SH/diag> ?
- Back to the root commands
ATA24-SH/diag> ..

A.4.2 Learning_table Commands

- Help
ATA24-SH/diag> learning_table ?
- Learning_table commands usage
ATA24-SH/diag> Learning_table

A.4.3 Netstat Commands

- Help
ATA24-SH/diag> netstat ?
- Netstat commands usage
ATA24-SH/diag> netstat -h
- Netstat diagnostics utility
ATA24-SH/diag> netstat <cmd>

A.4.4 Nslookup Commands

- Help
ATA24-SH/diag> nslookup ?
- Nslookup diagnostics utility
ATA24-SH/diag> nslookup <IPorDomainName>

A.4.5 Ping Commands

- Help
ATA24-SH/diag> ping ?
- Ping commands usage
ATA24-SH/diag> ping

- Ping diagnostics utility
ATA24-SH/diag> ping <Source Interface> <Destination Address>

<Source Interface>	0 : LAN 1 : WAN1 2 : WAN2 3 : WAN3
<Destination Address>	Domain name or IP Address of destination

A.4.6 Traceroute Commands

- Help
ATA24-SH/diag> traceroute ?
- Display usage message
ATA24-SH/diag> traceroute
- Traceroute diagnostics utility
ATA24-SH/diag> traceroute <cmd>

<cmd>	Octet string
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A.5 Firewall Commands

A.5.1 General Commands

- Enter firewall configuration function
ATA24-SH>firewall
- Help in the firewall function
ATA24-SH/ firewall > ?
- Back to the root commands
ATA24-SH/ firewall > ..

A.5.2 DoS Commands

- Help
ATA24-SH/ firewall >dos ?
- Set the icmpflood detection function
ATA24-SH/ firewall >dos /icmpflood
- Set the packet block detection function
ATA24-SH/ firewall >dos/packetblock
- Set the port scan detection function
ATA24-SH/ firewall >dos/ portscan
- Set the synflood detection function
ATA24-SH/ firewall >dos/ synflood
- Set the udpflood detection function
ATA24-SH/ firewall >dos/ udpflood
- Enable Dos Command
ATA24-SH/ firewall >dos/enable

A.5.2.1 Icmpflood Command

- Help
ATA24-SH/ firewall >dos >icmpflood ?
- Icmpflood commands usage

ATA24-SH/firewall/dos/icmpflood> enable <Option>

ATA24-SH/firewall/dos/icmpflood>threshold<Value> <Timeout>

<Option>	0: disable ICMPFlood detection function 1: enable ICMPFlood detection function
<Value>	0-65535, default=300 packets/sec
<Timeout>	The value of time out

A.5.2.2 Packetblock Command

- Help

ATA24-SH/network>dos >packetblock ?

- Packetblock commands usage

ATA24-SH/firewall/dos/packetblock > option <Value>

<Value>	1: Enable block ip option 2: Enable block TCP option 4: Enable block land 8: Enable tear drop 16:Enable block smurf 32:Enable block ping of death 64:Enable block trace route 128:Enable block icmp fragement 256:Enable SYN fragement 512:Enable Unknow protocol 1024:Enable Fraggie attrack
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A.5.2.3 Portscan Command

- Help

ATA24-SH/network>dos >portscan ?

- Portscan commands usage

ATA24-SH/firewall/dos/portscan > enable <Option>

ATA24-SH/firewall/dos/portscan > threshold <Value>

<Option>	0: disable port scan detection function 1: enable port scan detection function
<Value>	0-65535, default=300 packets/sec

A.5.2.4 Synflood Command

- Help

ATA24-SH/network>dos >synflood ?

- Portscan commands usage

ATA24-SH/firewall/dos/synflood >enable <Option>

ATA24-SH/firewall/dos/synflood >threshold <Value>

<Option>	0: disable SynFlood detection function 1: enable SynFlood detection function
<Value>	0-65535, default=300 packets/sec

<Timeout>	The value of time out
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A.5.2.5 Udpflood Command

- Help
ATA24-SH/network>dos >udpflood ?
- Portscan commands usage
ATA24-SH/firewall/dos/udpflood >enable <Option>
ATA24-SH/firewall/dos/udpflood >threshold <Value>

<Option>	0: disable UDPFlood detection function 1: enable UDPFlood detection function
<Value>	0-65535, default=300 packets/sec
<Timeout>	The value of time out

A.5.2.6 Enable Command

- Help
ATA24-SH/network>dos >enable ?
- Portscan commands usage
ATA24-SH/firewall/dos >enable <Option>

<Option>	0: disable DoS Function 1: enable DoS Function
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A.6 Network Commands

A.5.1 General Commands

- Enter network configuration function
ATA24-SH> network
- Help in the network diagnostics function
ATA24-SH/network> ?
- Back to the root commands
ATA24-SH/network> ..

A.5.2 MGN Commands

- Help
ATA24-SH/network>mgn ?
- Set the dhcp mode
ATA24-SH/network/ mgn > dhcp
- Set the IP address
ATA24-SH/network/ mgn > ip
- Set the Mac Address Manually
ATA24-SH/network/ mgn > mac

A.5.2.1 DHCP Command

- Help
ATA24-SH/network/ mgn >dhcp ?

- Display DHCP setting
ATA24-SH/network/ mgn >dhcp -s
- Enable/disable LAN setting
dhcp -mode <Index> <Mode>
- Specify range for LAN IP address
dhcp -range <Index> <Start IP> <End IP>
- Specify DNS server
dhcp -dns <Index> <Primary DNS> <Secondary DNS>
dhcp -dns <Index> <Primary DNS>
- Specify gateway
dhcp -gateway <Index> <Gateway IP>
- Specify lease time
dhcp -lease <Index> <Lease Time>
- Specify DHCP server
dhcp -relay <WAN IF> <DHCP Server IP>

<Mode>	0: Disable 1: Enable 2: Relay Agent
<Start IP>	IP address as starting point.
<End IP>	IP address as ending point.
<Primary DNS>	IP address as primary DNS.
<Secondary DNS>	IP address as secondary DNS.
<Gateway IP>	IP address as gateway.
<Lease Time>	Unit is minute.

A.5.2.2 IP Command

- Help
ATA24-SH/network/ mgn >ip ?
- Display nat setting
ATA24-SH/network/ mgn >ip -s
- Edit IP_NAT setting
ATA24-SH/network/ mgn >ip -w <IP> <Netmask>

<IP>	IP address.
<Netmask>	Subnet mask for NAT.

A.5.2.3 MAC Command

- Help
ATA24-SH/network/mgn>mac ?
- Display IP route setting
ATA24-SH/network/mgn>mac 1 <Mac Address>

< Mac Address >	xx:xx:xx:xx:xx:xx
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A.5.3 WAN Commands

- Help
ATA24-SH/network/wan ?

A.5.3.1 Backup Configuration Command

- Help
ATA24-SH/network/wan>advance> backup?
- Display the setting
ATA24-SH/network/wan>advance> backup -s
- Edit the setting
ATA24-SH/network/wan>advance> backup <status>

<status>	0: Disable 1: Enable
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A.5.3.2 Set WAN to Active Command

- Help
ATA24-SH/network/wan>active ?
- Edit WAN setting
ATA24-SH/network/wan>active <index> <status> <default route>
ATA24-SH/network/wan>active <index> <status> <default route>
<loadbalance><backupmaster> <backupslave>

<index>	1: WAN1 2: WAN2
<status>	0: not active 1: active
<default route>	0: not default 1: default
<loadbalance>	0: not join loadbalance 1: join loadbalance
<backupmaster>	0: not backupmaster 1: backupmaster
<backupslave>	0: not backupslave 1: backupslave

A.5.3.3 Set WAN to DHCP Mode Command

- Help
ATA24-SH/network/wan>dhcp ?
- Display current setting
ATA24-SH/network/wan>dhcp -s <index>
- Edit WAN setting
ATA24-SH/network/wan>dhcp <index>
ATA24-SH/network/wan>dhcp <index> <hostname> <domainname>

<index>	1: WAN1 2: WAN2
< hostname >	Name of the host.
< domainname >	Name of the domain

A.5.3.4 Configure MAC Address Command

- Help
ATA24-SH/network/wan>mac ?
- Display current setting
ATA24-SH/network/wan>mac -s <index>
- Edit WAN setting
ATA24-SH/network/wan>mac <index> <Use Default>
ATA24-SH/network/wan>mac <index> <User Define> <Mac Address>

<index>	1: WAN1 2: WAN2
<Use Default>	0: use default setting
<User Define>	1: user defined setting
<Mac Address>	MAC address for user defined configuration

A.5.3.8 Configure UP/Downstream Rate Command

- Help
ATA24-SH/network/wan>rate ?
- Display current setting
ATA24-SH/network/wan> rate -s <index>
- Edit WAN setting
ATA24-SH/network/wan> rate <index> <Downstream> <Upstream>

<index>	1: WAN1 2: WAN2
<Downstream>	0: using default setting (102400) Type any number to set downstream rate.
<Upstream>	0: using default setting(102400) Type any number to set upstream rate.

A.5.3.9 Show WAN Configuration Command

- Help
ATA24-SH/network/wan>show ?
- Display all WAN interfaces settings
ATA24-SH/network/wan> show
- Display specified WAN interface settings
ATA24-SH/network/wan>show <index>

<index>	1: WAN1 2: WAN2
<index>	1: WAN1 2: WAN2

A.5.3.10 Configure WAN Speed Command

- Help
ATA24-SH/network/wan>speed ?
- Display current setting

ATA24-SH/network/wan> speed -s <index>

- Edit WAN setting

ATA24-SH/network/wan>speed <index> <Speed & Duplex>

<index>	1: WAN1 2: WAN2
<Speed & Duplex>	1:Auto Negotiation 2:100M / Full Duplex 3:100M / Half Duplex 4:10M / Full Duplex 5:10M / Half Duplex

A.5.3.11 Set WAN to Static Mode Command

- Help

ATA24-SH/network/wan>static ?

- Display current setting

ATA24-SH/network/wan> static -s <index>

- Edit WAN setting

**ATA24-SH/network/wan> static <index> <IP> <Netmask> <Gateway>
<Primary DNS> <Secondary DNS>**

<index>	1: WAN1 2: WAN2
<IP>	Private IP address for WAN.
<Netmask>	Subnet mask for WAN.
<Gateway>	Private IP address for gateway.
<Primary DNS>	Private IP address as primary DNS.
<Secondary DNS>	Private IP address as secondary DNS.

A.5.3.12 Static Connection Detection Command

- Help

ATA24-SH/network/wan>static_detect ?

- Display current setting

ATA24-SH/network/wan> static_detect -s <index>

- Set condition for detection, sending ARP to Gateway

**ATA24-SH/network/wan> static_detect <index> 0 <detect interval>
<No-Reply Count>**

- Set condition for detection, sending PING

**ATA24-SH/network/wan> static_detect <index> 1 <detect interval>
<No-Reply Count> <detect destination>**

- Set condition for detection, sending HTTP

**ATA24-SH/network/wan> static_detect <index> 2 <detect interval>
<No-Reply Count> <detect destination>**

<index>	1: WAN1 2: WAN2
<detect interval>	Assign a number as interval time for detecting.
<No-Reply Count>	Assign a number (times) to ensure the connection of the WAN is on. After passing the times you set in this field and no reply received by the adapter, the connection of WAN

	interface will be regarded as breaking down.
<detect destination>	Private IP address or domain name

A.6 System Commands

A.6.1 General Commands

- Enter system configuration function
ATA24-SH> system
- Help in the system configuration function
ATA24-SH/system> ?
- Back to the root commands
ATA24-SH/system> ..

A.6.2 View ARP Cache Table Command

- Help
ATA24-SH/system/DiagnosticTools> arpcachetable ?
- Display the setting
ATA24-SH/system/DiagnosticTools> arp cache table

A.6.3 View DHCP Assignment Command

- Help
ATA24-SH/system/DiagnosticTools> dhcpassignmenttable ?
- Display the setting
ATA24-SH/system/DiagnosticTools> dhcp assignment table

A.6.4 View Routing Table Command

- Help
ATA24-SH/system/DiagnosticTools> routingtable ?
- Display the setting
ATA24-SH/system/DiagnosticTools> routing table

A.6.5 Administrator Control Commands

- Help
ATA24-SH/system> administrator ?
- Edit password for administrator
ATA24-SH/system> administrator<old password> <new password>
<verify password>

<old password>	Type old password.
<new password>	Type new password.
<verify password>	Retype the password for verification.

A.6.6 Auto Logout Commands

- Help
ATA24-SH/system > auto_logout ?

- Display the setting
ATA24-SH/system > auto_logout -s
- Edit the max-cli-session number
ATA24-SH/system > auto_logout -n <MaxSess>
- Kill the #'s log-session
ATA24-SH/system > auto_logout -d <SessNum>
- Edit the maximum idle time of auto logout
ATA24-SH/system > auto_logout -m <MaxIdleTime>
- Enable/Disable the auto logout
ATA24-SH/system > auto_logout <Active>

<MaxSess>	Integer(1 to15)
<SessNum>	Integer(1 to MaxSess)
<MaxIdleTime>	Seconds, Integer(10 to 86400)
<Active>	0: Disable 1: Enable

A.6.7 Config Commands

- Help
ATA24-SH/system> config ?
- Display the setting
ATA24-SH/system> config -s
- Execute the backup action
ATA24-SH/system> config backup <fname> <servIP>
- Execute the restore action
ATA24-SH/system> config restore <fname> <servIP>

<fname>	Octets string maximum length is 64.
<servIP>	IP address for the ATA-24 SH

A.6.8 Manage Port Commands

- Help
ATA24-SH/system> manage_port ?
- Display the setting
ATA24-SH/system> manage_port -s
- Manage port from WAN interface
ATA24-SH/system> manage_port -m <Use Default Port or Not><Manage from WAN>
- Reboot the system to apply the changes
ATA24-SH/system> manage_port -r
- Enable HTTP/Telnet function
ATA24-SH/system> manage_port -e <HTTP Enable> <TELNET Enable>
- Change port number for HTTP/Telnet function
ATA24-SH/system> manage_port -p <Http> <Telnet>
- Set IP address for the connection through WAN interface
ATA24-SH/system> manage_port -i <index> <IP Start> <IP End>

<Use Default Port or Not>	0 : Default 1 : User Define
<Http>	default: 80
<Telnet>	default: 23
<Manage from WAN>	0 : Disable all from Wan; 1 : Enable all from Wan; 2 : Enable only defined Wan IP;
<IP Start>	Starting point
<IP End>	Ending point.

A.6.9 Reboot Commands

- Help
ATA24-SH/system> reboot ?
- Reboot the system
ATA24-SH/system> reboot
- Reboot the system with keeping some important configuration
ATA24-SH/system> reboot keep
- Reboot the system with factory default configuration
ATA24-SH/system> reboot default
- Reboot the ATA-24 VoIP board only
ATA24-SH/system> reboot voip

A.6.10 Show Status Command

- Help
ATA24-SH/system> status ?
- Display the system status
ATA24-SH/system> status

A.6.11 Syslogd Commands

- Help
ATA24-SH/system> syslogd ?
- Display the syslog setting
ATA24-SH/system> syslogd -s
- Set IP address and port number for Syslog server
ATA24-SH/system>syslogd <Active> <RIP> <RPort> <Facility>
<Severity>

<Active>	0: Disable 1: Enable
<RIP>	Type IP address for LAN
<RPort>	Integer(1 to 65535)
<Facility>	0: local use 0 (local0)(default) 1: local use 1 (local1) 2: local use 2 (local2) 3: local use 3 (local3) 4: local use 4 (local4) 5: local use 5 (local5) 6: local use 6 (local6) 7: local use 7 (local7)

<Severity>	0: Emergency(default setting) 1: Alert 2: Critical 3: Error 4: Warning 5: Notice (including SIP) 6: Informational 7: Debug
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A.6.13 Upgrade Commands

- Help

ATA24-SH/system> upgrade ?

- Display the setting

ATA24-SH/system> upgrade -s

- Execute the firmware upgrade

ATA24-SH/system> upgrade <File Name> <Server IP>

<File Name>	Octets string maximum length is 64.
<Server IP>	Type IP address for the ATA-24 SH.

A.7 Voip Commands

A.7.1 General Commands

- Enter voip configuration function
ATA24-SH> voip
- Help in the voip diagnostics function
ATA24-SH/voip> ?
- Back to the root commands
ATA24-SH/voip> ..

A.7.2 H248 Commands

- Help
ATA24-SH/voip>h248 ?
- Display H248 call agent setting
ATA24-SH/voip/h248 > callagent -s
- Edit the H248 call agent setting
ATA24-SH/voip/h248>callagent <IPAddress> <Port>
- Display digit map default short/long timer setting
ATA24-SH/voip/h248 >dmTimer -s
- Edit the digit map timer setting
ATA24-SH/voip/h248>dmTimer <Termination> <Timer> <Sec>
- Display local listening port number for H248
ATA24-SH/voip/h248 >localport -s
- Edit the local listening port setting
ATA24-SH/voip/h248>localport <Port>
- Display message ID
ATA24-SH/voip/h248 >mid -s
- Edit message ID
ATA24-SH/voip/h248>mid -m <Mode>
ATA24-SH/voip/h248>mid -i <IP Mode>
ATA24-SH/voip/h248>mid <IPAddress>
ATA24-SH/voip/h248>mid <IPAddress> <Port>
- Display termination ID
ATA24-SH/voip/h248 >termid -s
- Edit termination ID
ATA24-SH/voip/h248>termId -a <Prefix> <StartNum>
ATA24-SH/voip/h248>termId <Termination> <ID>

<IPAddress>	Domain name or IP Address
<Port>	1 to 65535
<Termination>	1 to 24
<Timer>	0: short timer 1: long timer
<Sec>	1 to 99 (sec)
<Mode>	0: [IPAddress]:Port 1: [IPAddress]

<IP Mode>	0: WAN IPAddress 1: Manual IPAddress
<Prefix>	ID Name prefix
<StartNum>	Beginning of ID Name Number
<Termination>	1 to 24
<ID>	Identification name

A.7.3 Linetest Commands

- Help
ATA24-SH/voip>linetest ?
- Execute voip line card test
ATA24-SH/voip/linetest > line_card_test <Line> <TestItem>
- Execute voip metallic loop test
ATA24-SH/voip/linetest > metallic_loop_test <Line>
- Execute voip user phone test
ATA24-SH/voip/linetest >user_phone_test <Line> <TestItem>

<Line>	Available number: 1 to 24
<TestItem> (for voip line card test)	A: Normal Battery B: Loop Current C: Dial Tone Test D: Dial Digit Test E: Ring Voltage Test
<TestItem> (for voip user phone test)	A: DTMF Tone Testing B: Dial Pulse Testing C: Howler Tone D: Ringing

A.7.4 MGCP Commands

- Help
ATA24-SH/voip/mgcp> callagent ?
- Display the call agent setting
ATA24-SH/voip/mgcp> callagent -s
ATA24-SH/voip/mgcp> callagent2 -s
- Edit the IP address and port number for call agent
ATA24-SH/voip/mgcp> callagent <IPAddress> <Port>
ATA24-SH/voip/mgcp> callagent2 <IPAddress> <Port>
- Display the setting of End Point Name ID Style
ATA24-SH/voip/mgcp> epidstyle -s
- Edit the style mode for end point
ATA24-SH/voip/mgcp> epidstyle -m<Mode>
- Edit the logic ID for end point
ATA24-SH/voip/mgcp> epidstyle -l <LogicID>
- Edit the domain name for end point
ATA24-SH/voip/mgcp> epidstyle -d <DomainName>

- Display the MGCP heartbeat setting
ATA24-SH/voip/mgcp> **heartbeat -s**
- Edit the dual_homing action
ATA24-SH/voip/mgcp> **heartbeat <Active>**
- Edit the period of heartbeat for dual_homing
ATA24-SH/voip/mgcp> **heartbeat -t <Sec>**
- Edit the retry times of dual_homing
ATA24-SH/voip/mgcp> **heartbeat -r <Times>**
- Display local port setting
ATA24-SH/voip/mgcp> **localport -s**
- Edit the local port number for MGCP protocol
ATA24-SH/voip/mgcp> **localport <Port>**
- Display the port lock setting
ATA24-SH/voip/mgcp> **portlock -s**
ATA24-SH/voip/mgcp> **portlock -s <Port>**
- Edit the port lock/unlock
ATA24-SH/voip/mgcp> **portlock <Port> <lock>**
- Display the setting
ATA24-SH/voip/mgcp> **rsip -s**
- Set the RSIP action
ATA24-SH/voip/mgcp> **rsip <Active>**
- Display the setting pf sending RSIP with wildcarded endpoint ID
ATA24-SH/voip/mgcp> **rsip -s**
- Edit the RSIP action
ATA24-SH/voip/mgcp> **wildrsip <wildcard> <range>**

<IPAddress>	Assign an IP address of Call Agent server in MGCP (Default is 192.168.100.100)
<Port>	Assign a UDP port number to Call Agent server. 1 to 65535 (Default is 2727)
<Mode>	There are four options for users to select. (Default is 0) 0. aaln/#@[ip_addr] ex: aaln/1@[1.1.1.1] 1. mac_addr/#@[ip_addr] ex: 000504030201/1@[1.1.1.1] 2. aaln/#@mac_addr ex: aaln/1@000504030201 3. aaln/#@domain_name ex: aaln/1@callagent.com
<LogicID>	Starting number for logic ID.
<DomainName>	Name of the domain
<Active>	0: Disable 1: Enable (default=0) There are two options for users to select. Each endpoint sends its own RSIP Send only one wild-carded RSIP “Enable” to activate this function. “Disable” to close this function. (Default is Disable)

<Sec>	Integer(1 to 65535 default=60)
<Times>	Integer(1 to 300 default=1)
<Port> (for port lock/unlock)	1 to 24
<lock>	0: unlocked (default) 1: locked
<wildcard>	1: Enable wildcard(*) RSIP(Default) 0: Disable wildcard(*) RSIP
<range>	1: Enable range([1-24]) wildcards(Default) 0: Disable range([1-24]) wildcards

A.7.5 Miscellaneous Commands

- Help in the misc diagnostics function
ATA24-SH/voip>misc ?
- Display the dialing completion timeout
ATA24-SH/voip/misc> dialing_timeout -s
- Set the dialing completion timeout
ATA24-SH/voip/misc> dialing_timeout <value>
- Display echo cancellation configuration
ATA24-SH/voip/misc> echo_cancellation -s
- Enable echo cancellation configuration
ATA24-SH/voip/misc> echo_cancellation <enable>
ATA24-SH/voip/misc> echo_cancellation <enable> <tailLength>
- Display gain control setting
ATA24-SH/voip/misc>gain -s
- Set gain control setting
ATA24-SH/voip/misc>gain <Device port> <Speaker Gain> <Microphone Gain>
- Display line impedance parameter
ATA24-SH/voip/misc> lineimpedance -s
- Set same value for each line
ATA24-SH/voip/misc> lineimpedance <Country>
ATA24-SH/voip/misc>lineImpedance <line> <Country>
- Display line PCM codec
ATA24-SH/voip/misc> linepcmcodec -s
- Set same value for each line
ATA24-SH/voip/misc> linepcmcodec <codec>
ATA24-SH/voip/misc> linepcmcodec <line> c
- Display metering parameter
ATA24-SH/voip/misc> metering -s
- Set metering parameter
ATA24-SH/voip/misc> metering
ATA24-SH/voip/misc> metering -r <Reversal as Callee off-hook>
<Reversal as Callee on-hook>

- Display NAT traversal setting
ATA24-SH/voip/misc> nat -s
- Set NAT traversal setting
ATA24-SH/voip/misc>nat <Disable Mode>
ATA24-SH/voip/misc>nat <Manual Mode> <NatIpAddr>
ATA24-SH/voip/misc>nat <Auto Mode> <Type> <LocalPort>
<ServerIP> <ServerPort>
ATA24-SH/voip/misc>nat -sym <sym_rtp_t38>
- Display Line offhook detect current value
ATA24-SH/voip/misc>offhookdetect -s
- Set Line offhook detection
ATA24-SH/voip/misc> offhookdetect <Current>
ATA24-SH/voip/misc> offhookdetect <line> <Current>
- Display pulse timing configuration
ATA24-SH/voip/misc> pulsetime -s
- Set pulse timing
ATA24-SH/voip/misc> pulsetime <breakMin> <breakMax> <flashMin>
<flashMax> <makeMin> <makeMax> <interdigitmin>
- Display ring cadence and frequency setting
ATA24-SH/voip/misc> ring -s
- Set ring cadence and frequency (same value for each line)
ATA24-SH/voip/misc> ring -f <Frequency>
- Set ring frequency
ATA24-SH/voip/misc>ring -f <line> <Frequency>
- Set ring cadence
ATA24-SH/voip/misc>ring -c <Index> <Ton1> <Toff1> <Ton2> <Toff2>
<Ton3> <Toff3> <Ton4> <Toff4>
- Display the port number for sending/receiving RTP packets
ATA24-SH/voip/misc> rtp_port -s
- Set the port number for sending/receiving RTP packets
ATA24-SH/voip/misc> rtp_port <Port number>
- Display T.38 Fax Relay Configuration
ATA24-SH/voip/misc> t38 -s
- Set T.38 Fax Relay
ATA24-SH/voip/misc> t38 <Mode>
ATA24-SH/voip/misc t38 <Mode> <Port> <Redundancy>
- Display Voice Band Data (VBD) Configuration
ATA24-SH/voip/misc> vbd -s
- Set Voice Band Data (VBD) (same value for each line)
ATA24-SH/voip/misc> vbd <VBD>
- Set Voice Band Data (VBD)
ATA24-SH/voip/misc> vbd <port><VBD>

<value>	Range: 1~60 (second)
<enable>	0: disable

	1: enable
<tailLength>	Network Echo Canceller Tail Length (ms) Range: 8 ~ 128, should be multiple of 8
<Mode>	0: disable 1: enable
<Device port>	Device port number
<Speaker Gain>	Assign the gain value while receiving voice, default value is 0. The range is from -14 to 6.
<Microphone Gain>	Assign the gain value while transmitting voice, default value is 0. The range is from -14 to 6. (Default is 0)
<line>	Device line number (from 1 to 24)
<Country>	0: 600 Ohm (default) 1: 900 Ohm 2: China
<Codec>	0: Mu-LAW (default) 1: A-LAW
<Reversal as Callee off-hook>	0: Disable (default) 1: Enable
<Reversal as Callee on-hook>	0: Disable (default) 1: Enable
<Disable Mode>	0 : Disable NAT traversal (DEFAULT)
<Manual Mode>	1 : Manually input NAT IP address
<Auto Mode>	2 : Auto discover NAT IP address
<NatIpAddr>	Type IP address for manual mode.
<Type>	0 : Semi-auto, need to configure NAT 1 : Full-auto, no need to configure NAT
<LocalPort>	Local listening port number for STUN client
<ServerIP>	The IP address of STUN server
<ServerPort>	The port number of STUN server
<sym_rtp_t38>	0 : Disable symmetric RTP and T.38 1 : Enable symmetric RTP and T.38
<Current>	8: 8 mA (default) 10: 10 mA 12: 12 mA 15: 15 mA
<breakMin>	Minimum pulse break time (ms)
<breakMax>	Maximum pulse break time (ms)
<flashMin>	Minimum flash break time (ms)
<flashMax>	Maximum flash break time (ms)
<makeMin>	Minimum pulse make time (ms)
<makeMax>	Maximum pulse make time (ms)
<interDigitMin>	Minimum pulse inter digit time (ms)
<Frequency>	Ring frequency 20: 20 HZ (default) 25: 25 HZ
<Index>	Pattern Index, Index Value: 1-8
<Ton1>	Ton1 of cadence, unit: (ms)
<Toff1>	Toff1 of cadence, unit: (ms)
<Ton2>	Ton2 of cadence, unit: (ms)
<Toff2>	Toff2 of cadence, unit: (ms)

<Ton3>	Ton3 of cadence, unit: (ms)
<Toff3>	Toff3 of cadence, unit: (ms)
<Ton4>	Ton4 of cadence, unit: (ms)
<Toff4>	Toff4 of cadence, unit: (ms)
<Port number>	1 to 65535
<Mode>	0: Disable 1: Enable
<Port>	T.38 Starting Port, 1 to 65535 (default:13456)
<Redundancy>	T.38 Redundancy Number, 0 to 4 (default:1)
<port>	device port number
<VBD>	0: Auto Detection 1: Modem 2: Fax

Note: “Auto Discovery NAT IP Address” option is used when ATA-24 SH is behind a NAT adapter, NAT uses dynamic WAN IP address like as DHCP client. There must be having a STUN server in Internet. ATA-24 SH needs to negotiate with STUN server for this function.

Note: The “STUN”(Simple Traversal of UDP through NATs) server is an implementation of the STUN protocol that enables STUN functionality in SIP-based systems. STUN is an application-layer protocol that can determine the public IP and nature of a NAT device that sits between the STUN client and STUN server.

A.7.6 SIP Commands

- Help in the sip configuration function
ATA24-SH/voip/sip> ?
- Enter incallbarring configuration function
ATA24-SH/voip/sip> incallbarring
- Display allow list of incoming calls (for SIP)
ATA24-SH/voip/misc> allow -s
ATA24-SH/voip/misc>allow -s <Index>
- Edit allow list of incoming calls (for SIP)
ATA24-SH/voip/sip> allow -e <Index> <Name> <IP/Domain>
- Delete allow list of incoming calls (for SIP)
ATA24-SH/voip/sip> allow -d <Index>
ATA24-SH/voip/sip> allow -d
- Display deny list of incoming calls (for SIP)
ATA24-SH/voip/misc> deny -s
ATA24-SH/voip/misc> deny -s <Index>
- Edit deny list of incoming calls (for SIP)
ATA24-SH/voip/sip> deny -e <Index> <Name> <IP/Domain>
- Delete deny list of incoming calls (for SIP)
ATA24-SH/voip/sip> deny -d <Index>
ATA24-SH/voip/sip> deny -d

- Display current settings for incoming call barring (for SIP)
ATA24-SH/voip/misc> set -s
- Edit deny list of incoming calls (for SIP)
**ATA24-SH/voip/sip> set <Class> <MatchName> <MatchIP>
<SpeeddialFrom> <SpeeddialTo>**
- Display call waiting setting
ATA24-SH/voip/misc> callwait -s
- Edit call waiting setting
ATA24-SH/voip/sip>callwait <Port> <Mode>
- Display the codec setting
ATA24-SH/voip/sip> codec -s
- Edit perfect codec, codec rate and VAD for the port#
ATA24-SH/voip/sip> codec <Port> <PreferCodec> <CodecRate> <VAD>
- Edit single codec for the port#
ATA24-SH/voip/sip> codec -single <Port> <Active>
- Display VoIP setting
ATA24-SH/voip/sip>default_account -s
- Edit default SIP account
ATA24-SH/voip/sip>default_account <Port> <SIP Account>
- Display dialplan setting
ATA24-SH/voip/sip>dialplan -s
- Display dialplan setting with detail description
ATA24-SH/voip/sip>dialplan -h
- Edit dialplan setting (adding new entry)
**ATA24-SH/voip/sip>dialplan -a <MatchString> <MinLength>
<MaxLength> <PrefixStrip> <PrefixAdd> <SipIpAddr>
<InterDigitTimeOut> <Memo>**
- Edit dialplan setting (modifying an entry)
**ATA24-SH/voip/sip>dialplan -e <EntryIdx> <MatchString>
<MinLength> <MaxLength><PrefixStrip> <PrefixAdd> <SipIpAddr>
<InterDigitTimeOut> <Memo>**
- Delete dialplan setting
**ATA24-SH/voip/sip>dialplan -d <EntryIdx>
ATA24-SH/voip/sip>dialplan -D**
- Display DTMF Relay setting
ATA24-SH/voip/sip>dtmf_relay -s
- Edit DTMF relay mode for the port#
ATA24-SH/voip/sip>dtmf_relay <Port> <Mode>
- Edit DTMF relay mode and SIP INFO mode for the port#
**ATA24-SH/voip/sip>dtmf_relay <Port> <Mode> <SipInfoMode>
ATA24-SH/voip/sip>dtmf_relay -gain <port> <Gain Value>**
- Display fax transporting setting
ATA24-SH/voip/sip> fax -s
- Edit fax mode for the port#
ATA24-SH/voip/sip> fax <Port> <Mode>

- Display hotline setting
ATA24-SH/voip/sip> hotline -s
- Enable/Disable the hotline function
ATA24-SH/voip/sip> hotline <Port> <Active>
- Edit the hotline number
ATA24-SH/voip/sip> hotline <Port> <Active> <<Digits>
- Display local listening port number for SIP
ATA24-SH/voip/sip> localport -s
- Edit SIP local port number
ATA24-SH/voip/sip> localport <Port>
- Display port activation setting
ATA24-SH/voip/sip> port_active -s
- Choose proxy for the port
ATA24-SH/voip/sip> port_active <Port> <Active>
- Display proxy server setting
ATA24-SH/voip/sip> server -s
- Enable/Disable the proxy server
ATA24-SH/voip/sip> server <Proxy#> <Active>
- Enable/Disable the proxy server and outbound proxy
ATA24-SH/voip/sip> server <Proxy#> <Active> <Outbound>
- Edit the proxy server parameters
**ATA24-SH/voip/sip> server <Proxy#> <Active> <Outbound>
<ProxyName> <ProxyIP> <ProxyPort> <RegistrarIP> <RegistrarPort>
<Expires> <Domain>**
- Display SIP message (for SIP)
**ATA24-SH/voip/sip> siplog <Mode>
ATA24-SH/voip/sip> siplog <Mode><Line>**
- Display SIP user agent setting
ATA24-SH/voip/sip> sipua -s <Index>
- Display ring port setting
ATA24-SH/voip/sip> sipua -r
- Edit SIP user agent setting
**ATA24-SH/voip/sip> sipua -e <Index> <Active> <UserName>
<Password> <DisplayName> <AuthId><CallForwardMode>
<CallForwardUrl> <CallForwardRing><Proxy> <CallNoRegister>
<RingType> <IpBind>**
- Edit ring port setting
ATA24-SH/voip/sip> sipua -r <Index> <RingPort> <Mode>
- Delete SIP user agent setting
**ATA24-SH/voip/sip> sipua -e
ATA24-SH/voip/sip> dialplan -D**
- Display speed dial setting
**ATA24-SH/voip/sip> speeddial -s
ATA24-SH/voip/sip> speeddial -s <start> <end>**
- Add speed dial number and destination for the entry
ATA24-SH/voip/sip> speeddial -a <Number> <Destination> <Memo>

- Edit speed dial number, destination and memo for the entry
**ATA24-SH/voip/sip> speeddial -e <Index> <Number> <Destination>
 <Memo>**
- Delete the entry of speed dial
ATA24-SH/voip/sip> speeddial -d <Index>
- Delete all entries of speed dial
ATA24-SH/voip/sip> speeddial -D
- Display ports that unlocked
ATA24-SH/voip/sip> unlock -s
ATA24-SH/voip/sip> unlock -s <Port>
- Execute port unlock
ATA24-SH/voip/sip> unlock <Port>

<Index>	1 to 30 1 to 32 for SIP user agent
<Name>	Name of the incoming calls
<IP/Domain>	IP address or domain name
<Class>	0 : Allow all incoming calls 1 : Allow only calls from allow list 2 : Allow only calls from speed dial entries 3 : Deny only calls from deny list 4 : Deny all incoming calls
<MatchName>	0 : Disable ; 1 : Enable
<MatchIP>	0 : Disable ; 1 : Enable
<SpeeddialFrom>	1 to 150
<SpeeddialTo>	1 to 150
<Port> <	Port number of the device. From 1 to 24
<Mode> for ring port setting/RTP threshold setting	0 : Disable ; 1 : Enable
<PreferCodec>	Select one Codec to be applied on this port. ATA-24 SH supports five Codecs. 0: G.711U(PCMU) -64kbps 1: G.711A(PCMA) -64kbps 2: G.729A -8kbps (Default is 2) 3: G.723.1 -6.3kbps 4: G.726-32kbps
<CodecRate>	Select one rate value to be applied on this port. 20/40 - for PCMU or PCMA (Default is 20) 20/40/60/80 - for G.729A (Default is 20) 30/60 - for G.723.1 (Default is 30) 20/40 - for G.726 (Default is 20)
<VAD>	“Enable” to activate VAD(Voice Activity Detection, also known as Silence Suppression) function. “Disable” to stop using VAD. (Default is Disable)
<Active>	“Enable” to activate this port. “Disable” to close this port. (Default is Disable)
<SIP Account>	1 to 32
<EntryIdx>	1 to 60
<MatchString>	Matched string, ex: 9011x.T, maximum 63

	characters.
<MinLength>	Min. length of digits, range: 0~63, default: 0 (only use for x.T (unfixed length))
<MaxLength>	Max. length of digits, range: 0~63, default:32
<PrefixStrip>	Number of prefix digits to strip, range: 0~63
<PrefixAdd>	Prefix string to be add, -1: none maximum 63 char.
<SipIpAddr>	SIP IP address or domain name, ex: iptel.org 0 for no specific address
<InterDigitTimeOut>	Override the inter-digits timeout, range: 1~60(sec) default: 4 (sec)
<Memo>	User-specified name for comment, maximum 63 characters. Users can add some descriptions for each number. (Default is none)
<Mode>	0: Disable 1: RFC2833 (Default is 1) 2: SIP INFO
<SipInfoMode>	Click one option to be applied in DTMF function. There are three options to be supported as below – Disable(Inband) RFC2833 SIP INFO 0: CISCO 1: NORTEL (If Mode is 1, default is none) (If Mode is 2, default is 0)
<Gain Value>	0 to 31
< Mode >	Select a mode to be applied on FAX function. There are two options to be supported as below – Transparent: FAX will be transmitted via voice channel, no fax relay nor Codec change will be involved. T.38 Relay: Using T.38 Fax Relay. It is the default value. 0: Transparent 1: T.38 Relay (Default is 1)
<Active >	0: Disable, 1: Enable Or 0: off, 1: on
<Digits >	Default is none
< Proxy#>	Proxy # is from 1 to 3.
< Outbound >	0: Disable (Default is 0) 1: Enable (It means that each SIP protocol packet will be sent to SIP proxy server always.)
< ProxyName >	Assign a name of SIP proxy server. (Default is none)
< ProxyIP >	Assign an IP address of SIP proxy server. (Default is 0)
< ProxyPort >	Assign a port number of SIP proxy server. 1...65535 (Default is 5060)
< RegistrarIP >	Assign an IP address or domain name of SIP

	register server. (Default is 0)
< RegistrarPort >	Assign a port number of SIP register server. 1...65535 (Default is 5060)
< Expires >	Assign a timeout value for SIP protocol, the default value is 300. (minimum 60 seconds)
<Domain>	Assign an IP address or domain name of SIP Domain/Realm. (Default is 0)
<Mode> for SIP Message	0: Output last 50 lines 1: Output last N lines
<Line> for SIP Message	Print last N lines for mode 1
<UserName>	SIP username
<Password>	SIP password
<DisplayName>	SIP display name
<AuthId>	SIP authentication ID
<CallForwardMode>	0: Disable 1: Call forwarding all calls 2: Call forwarding busy 3: Call forwarding no answer
<CallForwardUrl>	SIP url format, ex: 101@iptel.org
<CallForwardRing>	1~10 (rings)
<Proxy>	0: Don't use proxy server 1: use Proxy 1 2: use Proxy 2 3: use Proxy 3
<CallNoRegister>	0: Call with Registration 1: Call without Registration
<RingType>	0: Rings all ports in the group 1: Rings the first available port 2: Rings by round robin
<IpBind>	0: WAN 1: VPN/LAN1 2: VPN/LAN2 3: VPN/LAN3 4: VPN/LAN4
<RingPort>	1~24 port
<Index> for speed dial setting	1~150
<Number>	Assign a dialing phone number.Ex: 101
<Destination>	Assign an address of dialing destination. Ex: 101@iptel.org

A.7.7 Statistics Commands

- Help in the Statistics function
ATA24-SH/voip/Statistics > ?
- Display call statistics setting
ATA24-SH/voip/statistics> callstat
- Display the setting by port
ATA24-SH/voip/statistics> callstat <Port>

- Edit the range for callstat port
ATA24-SH/voip/statistics> callstat <Port> <Range>
- Display RTP statistics setting
ATA24-SH/voip/statistics> rtpstat
- Display the setting by port
ATA24-SH/voip/statistics> rtpstat <Port>
- Edit the range for rtpstat port
ATA24-SH/voip/statistics> rtpstat <Port> <Range>
- Display RTP threshold setting
ATA24-SH/voip/statistics> rtpthreshold -s
- Edit the value for rtpthreshold
**ATA24-SH/voip/statistics> rtpthreshold <mode> <delayLow>
<delayHigh> <jitterLow> <jitterHigh> <lostLow> <lostHigh> <timeout>**
- Display VoIP RTP alert setting
ATA24-SH/voip/statistics> showalert
- Display the setting by port
ATA24-SH/voip/statistics> showalert <Port>

<Port>	Port number of the device. From 1 to 24
<Range> for VoIP call statistics	0: 15 minutes 1: 24 hour
<delayLow> <delayHigh> <jitterLow> <jitterHigh> <lostLow> <lostHigh> <timeout>	Round Trip Delay Low Threshold (ms) Round Trip Delay High Threshold (ms) Jitter Low Threshold (ms) Jitter High Threshold (ms) Packet Loss Ratio Low Threshold (0..100%) Packet Loss Ratio High Threshold (0..100%) RTCP timeout (in seconds)
<Lowfreq>	Assign a low frequency number in Hertz unit. (unit is HZ) (Default is 350)
<Highfreq>	Assign a high frequency number in Hertz unit. (unit is HZ) (Default is 440)
<Ton1>	The duration of the first ringing. (10msec per unit) (Default is 0)
<Toff1>	The silence duration after the first ringing. (10msec per unit) (Default is 0)
<Ton2>	The duration of the next continuous ringing. (10msec per unit) (Default is 0)
<Toff2>	The silence duration after the next continuous ringing. (10msec per unit) (Default is 0)
<Type> for call ID setting	0: North America 1: JAPAN 2: ETSI (Default is 2) 3: DTMF

A.7.8 VoIP Status Commands

- Help in the Statistics function
ATA24-SH/voip/status> ?

- Display VoIP faults
ATA24-SH/voip/status>faultstatus
- Display VoIP FXS port hook state (onhook or offhook)
ATA24-SH/voip/status>hookstate
ATA24-SH/voip/status>hookstate<Port>
- Display VoIP connection Status
ATA24-SH/voip/status>portstatus
ATA24-SH/voip/status>portstatus <Port>
- Display VoIP SIP User Agent Registration Status
ATA24-SH/voip/status>sipuastatus
ATA24-SH/voip/status>sipuastatus <Port>
- Display VoIP Status
ATA24-SH/voip/status>voipstatus
ATA24-SH/voip/status>voipstatus <Mode>

<Port>	Port number of the device. From 1 to 24
<Mode>	0: disable 1: enable

A.7.9 Tone User Defined Commands

- Help in the Statistics function
ATA24-SH/voip/tone/user_defined> ?
- Display user defined tone setting
ATA24-SH/voip/tone/user_defined> busy -s
- Edit frequency and cadence for busy tone
ATA24-SH/voip/tone/user_defined> busy <Lowfreq> <Highfreq> <Ton1>
<Toff1> <Ton2> <Toff2>
- Display caller ID setting
ATA24-SH/voip/tone/user_defined> callerid -s
- Edit caller id type
ATA24-SH/voip/tone/user_defined> callerid <Type>
- Display the setting
ATA24-SH/voip/tone/user_defined> congestion -s
- Edit frequency and cadence for congestion tone
ATA24-SH/voip/tone/user_defined> congestion <Lowfreq> <Highfreq>
<Ton1> <Toff1> <Ton2> <Toff2>
- Display user defined dial tone setting
ATA24-SH/voip/tone/user_defined> dial -s
- Edit frequency and cadence for dial tone
ATA24-SH/voip/tone/user_defined> dial <Lowfreq> <Highfreq> <Ton1>
<Toff1> <Ton2> <Toff2>
- Display user defined ringing tone setting
ATA24-SH/voip/tone/user_defined> ringing -s
- Edit frequency and cadence for ringing tone
ATA24-SH/voip/tone/user_defined> ringing <Lowfreq> <Highfreq>
<Ton1> <Toff1> <Ton2> <Toff2>

- Display the country of the tone setting
ATA24-SH/voip/tone> region -s
- Choose the region for CPT setting
ATA24-SH/voip/tone> region <Region Number>
- Display CPT tone timer setting
ATA24-SH/voip/tone/ timer -s
- Edit CPT tone timer
ATA24-SH/voip/tone/timer <Tone> <Timer>

<Lowfreq> <Highfreq> <Ton1> <Toff1> <Ton2> <Toff2> for user defined ring tone	(unit is HZ) (Default is 440) (units is HZ) (Default is 480) (10msec per unit) (Default is 0) (10msec per unit) (Default is 0) (10msec per unit) (Default is 200) (10msec per unit) (Default is 400)
<Region Number>	Select one country area for using VoIP feature. There is one option User Defined for proprietary setting. 0 : User Defined 1 : Australia 2 : British (Default is 2) 3 : Canada 4 : China 5 : Denmark 6 : Finland 7 : France 8 : Germany 9 : Hong Kong 10 : India 11 : Japan 12 : Netherlands 13 : Norway 14 : Singapore 15 : Taiwan 16 : USA
<Tone> for CPT tone timer setting	1: Dial Tone 2: Busy Tone 3: Howler Tone 4: Ringing Tone 5: Special Dial Tone 6: Call waiting Tone 7: Congestion Tone 8: Reorder Tone
<Timer>	Range: 0~300 <sec>

A.7.10 Config Commands

- Help
ATA24-SH/voip>protocol ?

- Execute/activate VoIP setting
ATA24-SH/voip>config

A.7.11 List Commands

- Help
ATA24-SH/voip>listcmds ?
- Display all VoIP CLI commands
ATA24-SH/voip>listcmds

A.7.12 Protocol Commands

- Help
ATA24-SH/voip>protocol ?
- Display the setting
ATA24-SH/voip>protocol -s
- Set the voip protocol
ATA24-SH/voip>protocol <Protocol>

<Protocol>	0: MGCP, 1: SIP, 2:H.248
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