

# VigorTalk ATA-24 24-Port Analogue Terminal Adapter User's Guide

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The VigorTalk ATA-24 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 is depicted in Figure 1-1, which illustrates interconnections among branch offices through the Internet via the VigorTalk ATA-24 adapter. By combining with an existing PBX, an Internet phone from a remote branch can also access any extension number on a local PBX or a traditional phone via PSTN. 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. Combining a PBX with the adapter allows you to call anyone who has an Internet phone or a traditional telephone number – including local, long distance, mobile, and international numbers. Internet Telephony 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 has 3 WAN interfaces. This allows the system to reach peak performance and reduces the cost of maintaining a single high-speed trunk by sharing the load amongst the multiple WAN interfaces. Each interface can be connected to an individual Internet Service Provider. The VigorTalk ATA-24 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.

Te PWR VA	Re	LNK actory 1000 eset FDX LAN	1 2 3 4 5 6 7 8 9 10 11 12 13 14 15 16 9 10 11 12 13 14 15 16 1 2 3 17 18 19 20 21 22 23 24		
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.		
	LNK	On	The Ethernet link is established.		
LAN		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.		

LED		Status	Explanation
	LNK	On	The Ethernet link is established on corresponding port.
		Off	No Ethernet link is established.
WAN1/2/3	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.
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
Console	Provided for technician use.
LAN	Connecter for local networked devices.
WAN (WAN1 ~ WAN3)	Connecter for remote networked devices.

# **1.2 Hardware Installation**

Before starting to configure the adapter, you have to connect your devices correctly.

- Connect one end of an Ethernet cable (RJ-45) to one of the LAN ports of VigorTalk ATA-24. Connect the other end of the cable (RJ-45) to the Ethernet port on your computer (that device also can connect to other computers to form a small area network).
- 2. Connect a server/modem/adapter (depends on your requirement) to any available WAN port of the device with Ethernet cable (RJ-45).
- 3. Connect telephone sets to the **Phone** port of VigorTalk ATA-24 with telephone lines (RJ-11 to RJ-11).
- 4. Connect the **PSTN** port to PABX.
- 5. Connect the power cord to the power port of VigorTalk ATA-24 adapter on the rear panel, and the other side into a wall outlet.
- 6. Power on the device by pressing the power switch on the rear panel.
- 7. The system starts to initiate. The **PWR** LED should be **ON**.After completing the system test, the **ACT** LED will light up and start blinking. The **LAN/WAN** LED for that port on the front panel will light up.

Below shows an outline of the hardware installation for your reference (take VigorTalk ATA-24 as an example).



#### 1.2.1 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 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 theVigorTalk ATA-24, while the RS232 DB9 connects to a console port on the computer. The default setting of the console port is "**baud rate 57600, 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 autodetection of either straight or crossover RJ45 cables. These cables are used on WAN, LAN, and DMZ interfaces.



#### **Chassis Connections**

The VigorTalk ATA-24 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 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-24. 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 correctly.
  - Notice: You may either simply set up your computer to get IP dynamically from the adapter or 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.



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

System - Status		
Refresh Option:	No Refresh V Refresh	
Basic Status	LAN Status WAN Status	
Model :	ATA24 System	
Hardware Version :	0	
Firmware Version :	ATA24_TEST_V1.0.0.m825.RC17	
Build Date&Time :	Thu Jun 26 11:50:22 CST 2008	
System Uptime :	0 days 0 hours 7 minutes 24 seconds	
CPU Usage :	0.0000%	
Memory Size :	128 MBytes	
Memory Usage :	32.9430%	
Current System Time :	1970-01-01 00:07:24	
	DrayTek Corp. © 1997	- 2008 All rights reserved. DrayTek Enterprise Network Soluti

4. Go to System page and choose Change Password.

uick Setup System	Network Advan	ed Firewall	VoIP 15:26:2
Status			
Suptom Kine			
System Syslog			
Access Com	trol		1
Refresh Optio 🧟 Change Pass	sword	Refresh	
Basic St		/AN Status	
A niniware op	Jgraue I	AN Status	
Model : 🤯 Reboot	System		
Hardware Ver 🔝 Diagnostic T			
Firmware Version :	ATA24_TEST_V1.0		
Build Date&Time :	Thu Jun 26 11:50:2	2 CST 2008	
System Uptime :	0 days 0 hours 7 m	inutes 24 seconds	
CPU Usage :	0.0000%		
Memory Size :	128 MBytes		
Memory Usage :	32.9430%		
Current System Time :	1970-01-01 00:07:2	24	
			DrayTek Corp. © 1997 - 2008 All rights reserved. DrayTek Enterprise Network Solu

5. The following screen will appear.

System - Char	nge Password						
Old Password :							
New Password :	••••						
Confirm Password :	••••						
						Annha	Canaal
						Apply	Cancel
		Dr	ıyTek Corp. © 1997	- 2008 All rights res	served, DrayTek I	Enterprise Net	work Solution
		Dr	ıyTek Corp. © 1997	- 2008 All rights re:	served. DrayTek I	Enterprise Net	work Solution
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		Dr	y⊤ek Corp. © 1997	- 2008 All rights res	served. DrayTek I	Enterprise Net	work Solutio

- 6. 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.
- 7. Now, the password has been changed. Next time, use the new password to access the Web Configurator for this adapter.

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

Connect to 192.16	8.1.1 🛛 🛛 🔀
	GET
Login to ATA24	
User name:	🔮 admin 💌
Password:	••••
	Remember my password
	OK Cancel

# 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, DHCP, PPPoE, or PPTP 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.

IAC Address :	💿 Default MAC i O User	Defined MAC
	00:50:7f:2f:c4:c6	
)ownstream Rate :	102400	(kbps)
Jpstream Rate :	102400	(kbps)
ype :	Fast Ethernet 💌	
Physical Mode :	Auto Negotiation	

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 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 outgoing buffer adjustment. If you use a DSL subscriber service with a 256Kbps downstream, please set the downstream rate setting with 256Kbps.
Туре	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 <b>auto negotiation</b> , <b>full duplex</b> , and <b>half duplex</b> of either 10/100/1000M speed options for the WAN Interface.
IP Mode	Select an IP mode for this WAN interface. There are four available modes for Internet access, <b>Static</b> , <b>DHCP</b> , <b>PPPoE</b> , and <b>PPTP</b> . On this page you may configure the WAN interface to use <b>Static</b> (fixed IP), <b>DHCP</b> (dynamic IP address), <b>PPPoE</b> or <b>PPTP</b> . Most of the cable users will use the <b>DHCP</b> 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:

Static/DHCP Configuration	PPPoE/PPTP Configuration		
IP Address :	172.16.3.229	Host Name :	
Subnet Mask :	255.255.255.0	Domain Name :	
Default Gateway :	172.16.3.1	(Host Name and Domain	Name are required for some ISPs.)
Primary DNS :	168.95.1.1	]	
Secondary DNS :	168.95.192.1	]	
IP Alias List			
1.	10.1.1.100	2.	10.1.1.102
3.	10.1.1.103	4.	
5.		6.	
7.		8.	
			Next >>

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 LAN interface continuously.

#### LAN – LAN IP/DHCP Page

LAN IP/DHCP	
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 :	
Secondary DNS :	
Lease Time (Min) :	1440
Gateway IP(Optional) :	

IP Address	Assign an IP address for the LAN interface.
Subnet Mask	Assign the subnet mask for the LAN interface.
Status	Click <b>Enable</b> to use DHCP server; click <b>Disable</b> 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 automatically. It is not necessary for you to assign any setting. (Host Name and Domain Name are required for some ISPs). S imply click **Next** to setup LAN interface.

Quick Setup -	WAN
MAC Address :	Default MAC User Defined MAC     00:50:7f:2f:c4:c6
Downstream Rate :	102400 (kbps)
Upstream Rate :	102400 (kbps)
Type :	Fast Ethernet 💌
Physical Mode :	Auto Negotiation 🔽
IP Mode :	O Static ⊙ DHCP O PPPoE O PPTP
Static/DHCP Configuration	PPPoE/PPTP Configuration

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

#### LAN – LAN IP/DHCP Page

LAN IP/DHCP	
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 :	
Secondary DNS :	
Lease Time (Min) :	1440
Gateway IP(Optional) :	

IP Address	Assign an IP address for the LAN interface.
Subnet Mask	Assign the subnet mask for the LAN interface.
Status	Click <b>Enable</b> to use DHCP server; click <b>Disable</b> to close DHCP server; click <b>Relay Agent</b> 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.

#### 2.2.4 PPPoE

This mode is used for most of DSL modem users. All local users can share one PPPoE connection to access the Internet. Your service provider will give you the user name, password, and authentication mode for PPPoE settings.

Quick Setup - W	/AN
MAC Address :	Default MAC User Defined MAC     00:00:00:00:02
Downstream Rate :	102400 (kbps)
Upstream Rate :	102400 (kbps)
Туре:	Fast Ethernet 💌
Physical Mode :	Auto Negotiation 💌
IP Mode :	O Static O DHCP ⊙ PPPoE O PPTP

If your ISP provides you the **PPPoE** (Point-to-Point Protocol over Ethernet) connection, please select **PPPoE** for this adapter to get the following page. Enter the **username** and **password** provided by your ISP on the web page.

Configuration Conf User Name :	iguration 88991234@hinet.net	PPTP Local Address :	
Password :		PPTP Subnet Mask :	
assworu.	••••	FFIF Subilet Mask.	
Authentication :	PAP 💌	PPTP Server Address :	
Service Name (Optional):	hinet		

User Name	Assign a specific valid user name provided by the ISP.
Password	Assign a valid password provided by the ISP.
Authentication	Select <b>PAP</b> or <b>CHAP</b> protocol for PPP authentication. The default value is <b>PAP</b> .
Service Name	Assign a service name required from ISP service.

After setting up the WAN interface, the user can click Next to setup the LAN interface

#### continuously. LAN – LAN IP/DHCP Page

Quick Setup - LA	AN
LAN IP/DHCP	
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 :	
Secondary DNS :	
Lease Time (Min) :	1440
Gateway IP(Optional) :	

IP Address	Assign an IP address for the LAN interface.
Subnet Mask	Assign the subnet mask for the LAN interface.
Status	Click <b>Enable</b> to use DHCP server; click <b>Disable</b> to close DHCP server; click <b>Relay Agent</b> 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.

#### 2.2.5 PPTP

This mode lets user get the IP group information by a DSL modem with PPTP service from ISP. Your service provider will give you user name, password, and authentication mode for a PPTP setting.

Quick Setup - WAN		
MAC Address :	● Default MAC ● User Defined MAC 00:00:00:00:00:02	
Downstream Rate :	102400 (kbps)	
Upstream Rate :	102400 (kbps)	
Type :	Fast Ethernet 🛩	
Physical Mode :	Auto Negotiation 🗸	
IP Mode :		

If your ISP offers you **PPTP** (Point-to-Point Tunneling Protocol) mode, please select **PPTP** for this adapter. Next, enter the **PPTP Subnet Mask (e.g., 255.255.255.0), PPTP Local Address (e.g., 10.66.99.88)** and **PPTP Remote Address (e.g., 172.66.99.88)** provided by your ISP on the web page.

	E/PPTP guration		
User Name :	88991234@hinet.net	PPTP Local Address :	
Password :	••••	PPTP Subnet Mask :	
Authentication :	PAP 😽	PPTP Server Address :	
Service Name (Optional):			
			Next >>

**PPTP Local Address** Assign a local IP address of PPTP.

**PPTP Subnet Mask** Assign a net mask value for IP address of PPTP.

**PPTP Remote Address** Assign a remote IP address of PPTP server.

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

#### LAN – LAN IP/DHCP Page

Quick Setup - L	AN
LAN IP/DHCP	
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 :	
Secondary DNS :	
Lease Time (Min) :	1440
Gateway IP(Optional) :	

IP Address	Assign an IP address for the LAN interface.	
Subnet Mask	Assign the subnet mask for the LAN interface.	
Status	Click <b>Enable</b> to use DHCP server; click <b>Disable</b> to close DHCP server; click <b>Relay Agent</b> 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	<b>Option</b> 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.

Refresh Option:	No Refresh	
Basic Status L	AN Status WAN Status	
Model :	ATA24 System	
Hardware Version :	0	
Firmware Version :	ATA24_TEST_V1.0.0.m825.RC17	
Build Date&Time :	Thu Jun 26 11:50:22 CST 2008	
System Uptime :	0 days 1 hours 27 minutes 2 seconds	
CPU Usage :	0.9804%	
Memory Size :	128 MBytes	
Memory Usage :	32.9681%	
Current System Time :	1970-01-01 01:27:02	

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.	

#### LAN Status

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

Refresh Option:	No Refre	sh 🔽	Refresh
Basic Status	LAN Status	WAN Status	
IP Address :	192.168.1.1		
MAC Address :	00:50:7f:f0:4	4:f8	
High Availability Statu	s:		
RX Packets :	39024		
TX Packets :	40566		

IP Address	Displays the IP address of the LAN interface.
MAC Address	Displays the MAC address of the LAN Interface.
High Available Status	The High Available Status is shown when the function is enabled. When there are two VigorTalk ATA-24 devices in the same LAN, one can be set as Master device and the other can

	be set as Slave device.
	Master - It means that VigorTalk ATA-24 plays the Master role
	in high availability feature.
	Slave - It means that VigorTalk ATA-24 plays the Slave role in
	high availability feature.
	If there is only one VigorTalk ATA-24 used in LAN, this line
	will be blank.
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, DHCP, PPPoE, PPTP or DMZ) 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.

Refresh Option:	No Refresh 💌 Refresh		
Basic Status	LAN Status WAN Status		
WAN1 :		WAN2:	
IP Address :	172.16.3.229	IP Address :	
MAC Address :	00:50:7f:c5:45:89	MAC Address :	00:50:7f:c5:45:8a
Primary DNS :		Primary DNS :	
Secondary DNS :		Secondary DNS :	
Gateway:	172.16.3.4	Gateway:	
RX Packets :	1214754	RX Packets :	0
TX Packets :	35819	TX Packets :	0
Connection Status :	connected	Connection Status :	
Up Time :	0 days 22 hours 0 minutes 22 seconds	Up Time :	
	Disconnect		
WAN3 :			
IP Address :			
MAC Address :	00:50:7f;c5:45:8b		
Primary DNS :			
Secondary DNS :			
Gateway:			
RX Packets :	0		
TX Packets :	0		
Connection Status :			
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.
<b>Connection Status</b>	Displays the connection status of the WAN interface.
Up Time	Displays the total system uptime of the interface.
Disconnect	Disconnects current connection.

# 3.1.2 Time

As an NTP (Network Time Protocol) client, the adapter gets standard time from the time server. Some time-based functions, such as **Call Schedule** and **URL Content filtering**, cannot work properly until the system time functions run successfully. Typically, NTP achieves high accuracy and reliability with multiple redundant servers and diverse network paths.

The VigorTalk ATA-24 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	
OUse Browser Time	
NTP Server :	
Time Zone :	(GMT+00:00) Greenwich Mean Time : Dublin 💌
Daylight Saving Time :	⊙ Not Use O Use
Update Interval :	30 seconds V
	Apply Cancel

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 is located.
Daylight Savings Time	Select <b>Use</b> 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 <b>Apply</b> to save these settings.

# 3.1.3 Syslog

The VigorTalk ATA-24 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:

System - Syslog	
O Disable 💿 Enable	
Syslog Server IP :	0.0.0.0
Syslog Server Port :	514
	Apply Cancel

Status	Click <b>Enable</b> to activate this function. The adapter will send system log message for your reference. If you click <b>Disable</b> , 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 will not send Syslog packets to the Syslog server.
Syslog Server Port	Assign a port for the Syslog protocol.
Apply	Click <b>Apply</b> 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 with specified IP address.

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

System - A	Access Cont	rol	
Management Allow Management Management Allow Management O Disable	ent Method: Telnet Access Contro	SSH Enable User Defined Allowed IP1: Allowed IP2: Allowed IP3:	
Management	Port		
● Default Ports	(HTTP Port:80 Tell	net Port:23 SSH Port:22)	User Defined Ports HTTP Port: 80 Telnet Port: 23 SSH Port: 22
PING Restrict	ion		
🔲 Disable PIN	G from the LAN		
🔲 Disable PIN	G from the WAN		
			Apply Cancel

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	<b>Disable</b> - Disable the management from the WAN interface. <b>Enable All</b> - Enable all management (throughHTTP/Telnet/SSH) from the WAN interface.
	<ul><li>Enable User Defined WAN IP - System can be managed by these three IP addresses via WAN.</li><li>Allowed IP1(to 3) - Type in ranges for IP addresses (up to three) for managing the system.</li></ul>
Management Port	<ul><li>Default Ports - Use the default ports for HTTP and Telnet if you choose HTTP and Telnet as management methods.</li><li>User Defined Ports - Or you can assign new port numbers for HTTP, Telnet and SSH respectively.</li></ul>
PING Restriction	<b>Disable PING from the LAN -</b> Choose this function to reject all ICMP packets from LAN side. <b>Disable PING from the WAN -</b> 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 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.

System - Configuration		
Restore		
Select a configuration file:	Browse, Apply	
Backup		
Backup configuration file		
Push Backup button	Backup	

Select a Configuration FilePlease click the Browse... button to find out the location of the<br/>configuration file to be uploaded to the adapter and click<br/>Apply.Backup Configuration FileDownload the configuration file to a local host. The default<br/>file name is "ata24.cfg".

## 3.1.6 Firmware Upgrade

VigorTalk ATA-24 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) on the computer first.

System - Firmware Upgrade		
Caution :	After an upgrade procedure a reboot is required.	
Current Version :	ATA24 System ATA24_TEST_V1.0.0.m825.RC17	
Location :	⊙ Local O Remote	
Firmware :	Browse.	
TFTP Server IP		
Remote File Name	ATA24.all	
	Apply Cancel	

Caution	Displays a caution for your reference.
<b>Current Version</b>	Displays current firmware version that you are using.
Location	Local means upgrade firmware from browser.
	Remote 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 <b>Apply</b> 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 (<u>www.draytek.com.tw</u>) or FTP site (<u>ftp.draytek.com</u>) on your computer first.
- 2. Connect the RJ45 connector of console cable to the console port on VigorTalk ATA-24 and the DB9 connector of the console cable to the RS232 port on the PC.

System Prope	rties	×
General Net	work Identification Hardware User Profiles Advanced	
P F	vigor 3300 - HyperTerminal COM1 Properties	? ×
	Port Settings	
_	Bits per second: 115200	
	Data bits: 8	
	Parity: None	
	Stop bits: 1	
	Flow control: None	
	Restore Defaults	
	OK Cancel Apply	

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.

- 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.

```
File, Edit View Call Transfer Help
🗅 😹 🐵 🌫 🐨 🖆
  slot = 0 sector size
                                            65536
65536
65536
                                         .....
  slot - 0 sector size
  slot = 0 sector size =
  slot = 0 sector size =
                                            65536
  slot = 0 sector size = 65536
  Updating flash block at bfd30000
 set ethaddr0 00:50:7f:28:80:e3
set ethaddr1 00:50:7f:28:80:e4
set ethaddr2 00:50:7f:28:80:e4
 set #default_nif_wan1_mac 00:50:7f:28:80:e4
set #default_nif_wan2_mac 00:50:7f:28:80:e5
set #default_nif_wan3_mac 00:50:7f:28:80:e6
set #default_nif_wan4_mac_00:50:7f:28:80:e7
set flash0_0 "780000:80000:general"
 DrayTek Corporation VigorTalk ATA-24
(Firmware version: V2.5.7)
Hardware version: 0
 V3 board, for V3 GPIO config
have voip card
 Draytek login: VigorTalk ATA-24
 onnected 0:05:41
                         Auto detect
                                        57600 8-N-1
                                                                              N.N
```

#### 3.1.7 Reboot

The VigorTalk ATA-24 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 system has been rebooted.



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

System is rebo	oting, please wait
20	seconds left
If your current interface or management port configur	ation has been changed, please access with the new URL.

# 3.1.8 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 supports four functions, **Routing Table**, **ARP Cache Table**, **DHCP Assignment Table**, and **NAT Active Sessions 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:

Kernel IP rout				
Destination 192.168.1.0 172.16.0.0 127.0.0.0	0.0.0.0	255.255.0.0	Flags U U U	Interface vlan4040 vlan4041 lo

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 <b>vlan4040</b> if it is a LAN interface and <b>vlan4041</b> if it is a WAN interface.
Refresh	Click <b>Refresh</b> to re-display this web page for getting newest routing information.

• Select View ARP Cache Table to get the following page:

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

IP Address	Displays the IP address for different ARP cache.
MAC Address	Displays the MAC address for different ARP cache.
Interface	Denoted by <b>vlan4040</b> if it is a LAN interface. <b>vlan4041</b> means it is a WAN1 interface; <b>vlan4042</b> means it is a WAN2 interface; <b>vlan4043</b> means it is a WAN3 interface.
Refresh	Click <b>Refresh</b> to re-display this web page for getting newest ARP information.

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

168.1.10 00:0E:A	6:2A:D5:AI 23 hours,	. 52 minutes, 15	secor

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 <b>Refresh</b> 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 LAN interfaces for the adapter.



#### 3.2.1 WAN and Internet Access Setup

The VigorTalk ATA-24 supports three WAN interfaces (with four IP Modes - Static, DHCP, PPPoE, and PPTP), which share the same setting page. In the **Network** group, please click the **WAN** option. The following page will be shown.

Load Balanc	e:	💿 Disable i 🔘 E	nable ( 🗌 Auto W	eight)				
Backup :		💿 Disable 🔘 E	nable					
#	Edit	IP Mode	Active	Default Route	Load Balance	Weight	Backup-Master	Backup-Slave
WAN1	<b>E</b>	Static		۲		10% 👻		
WAN2	2	Not Set				10% 🗸		
MAN3	2	Not Set				10% 🗸		

Load Balance	Enables or disables the WAN load balance function. The <b>Auto Weight</b> option becomes available if <b>Enable</b> mode is selected. Load Balance allows the adapter distributing data in and out of the Internet by using different WAN interfaces at the same time.	
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 four options: Static, DHCP, PPPoE, PPTP, and DHCP.	
Active	Activates/closes this WAN interface.	
Default Route	Sets this WAN interface as default route interface.	
Load Balance	Adds this WAN interface to the load balance group.	
Weight	Sets the weight load (10-90%) for this WAN interface for load balance. This selection is available only when Auto Weight is unchecked.	
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 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 :	O Default MAC	
Downstream Rate :	102400 (kbps)	
Upstream Rate :	102400 (kbps)	
Physical Mode :	Auto Negotiation 💌	
IP Mode :		

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 <b>Static (fixed IP)</b> , <b>DHCP (dynamic IP address)</b> , <b>PPPoE</b> , or <b>PPTP</b> 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, you need to know what kind of Internet access your ISP provides. The following sections introduce four widely used broadband access services: **Static, PPPoE, PPTP** 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 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 ODHCP OPPOE OPI	PTP	
	DE/PPTP iguration		
IP Address :	172.16.3.229	Host Name :	
Subnet Mask:	255.255.0.0	Domain Name :	
Default Gateway :	172.16.3.4	(Host Name and Domain	Name are required for some ISPs.)
Primary DNS :			
Secondary DNS :			
MTU :	1500		
Connection Detectio	n		
Detect Type :	Send ARP to Gateway 💌		
Detect Interval(sec) :	10		
No-Reply Count:	2		
Detect Destination Host : (IP or Domain Name)			
IP Alias List			
1.		2.	
3.		4.	
5.		6.	
7.		8.	
<u>9-32</u>			
			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.
ΜΤυ	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 <b>Send ARP to Gateway</b> , <b>Send PING</b> and <b>Send HTTP Request</b> supported in ATA24.

	Send Http Request 👻 Send ARP to Gateway Send PING Send Http Request
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 <b>Detect Type</b> is set with <b>Send PING</b> or <b>Send Http</b> <b>Request</b> .
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 <b>Apply</b> to go back to the WAN Interface Configuration page. To apply all settings, click <b>Apply</b> 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 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 :	O Static ⊙ DHCP O PPPoE O PPTP
	E/PPTP guration
IP Address :	172.16.3.229 Host Name :
Subnet Mask :	255.255.0.0 Domain Name :
Default Gateway :	172.16.3.4 (Host Name and Domain Name are required for some ISPs.)
Primary DNS :	
Secondary DNS :	
MTU :	1500
Connection Detection	1
Detect Type :	Send ARP to Gateway 💌
Detect Interval(sec) :	10
No-Reply Count:	2
Detect Destination Host : (IP or Domain Name)	
	Apply Reset Cancel

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. Send Http Request Send ARP to Gateway Send PING Send Http Request	
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	Assign an IP address or Domain name as a destination to be	

(IP or Domain Name)	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 <b>Detect Type</b> is set with <b>Send PING</b> or <b>Send Http Request</b> .
Apply	Click <b>Apply</b> to go back to the WAN Interface Configuration page. To apply all settings, click <b>Apply</b> on the WAN Interface Configuration page and reboot your adapter.
Reset	Click this button to clear all the configurations for this page.

# **PPPoE** Configuration

Most DSL modem users will use this mode. All the local users can share one PPPoE connection to access the Internet.

IP Mode :	◯ Static ◯ DHCP 💽 PPoE ◯	РРТР	
	PPoE/PPTP nfiguration		
User Name :	88999	PPTP Local Address :	1.1.1.1
Password :	••••	PPTP Subnet Mask :	255.255.0.0
Authentication :	PAP 💌	PPTP Server Address :	1.1.1.3
Service Name :	hinet		
PPPoE IP Alias :	Enable		
MTU :	1442		
IP Address Assign	ment Method (IPCP)		
Fixed IP :	💿 No (Dynamic IP) 🔵 Yes		
Fixed IP Address :			
Connection Detec	tion		
Detect Interval :	10		
No-Reply Count :	2		
			Apply Reset Cancel

User Name	Assign a specific valid user name provided by local ISP.	
Password	Assign a valid password provided by local ISP.	
Authentication	Select <b>PAP</b> or <b>CHAP</b> protocol according to the feature that your ISP provided for widest compatibility. The default value is <b>PAP</b> . The password will be encrypted in CHAP but not in RAP.	
Service Name	Assign a service name required for some ISP services.	
<b>PPPoE IP Alias</b>	Enable IP Alias for PPPoE protocol.	
MTU	It means the maximum transmission unit. Default value is 1442. Change it if you want.	
Fixed IP	Click No to use dynamic IP for PPPoE protocol; click Yes to	

	use a fixed IP address. If you choose <b>Yes</b> , you have to type an IP address in the field of Fixed IP Address below.
Fixed IP Address	Type an IP address here if you click Yes as Fixed IP.
Detect Interval	Assign an interval time for detecting if the WAN connection is on or off.
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.
Apply	Click <b>Apply</b> to go back to the WAN Interface Configuration page. To apply all settings, click <b>Apply</b> on the WAN Interface Configuration page and reboot your adapter.
Reset	Click this button to clear all the configurations for this page.

# **PPTP Configuration**

The service provider must provide the exact settings for this mode.

IP Mode :	OStatic ODHCP OPPOE OPPT	P	
	E/PPTP guration		
User Name :	88999	PPTP Local Address :	1.1.1.1
Password :	••••	PPTP Subnet Mask :	255.255.0.0
Authentication :	PAP 🗸	PPTP Server Address :	1.1.1.3
Service Name :	hinet		
PPPoE IP Alias :	Enable		
MTU :	1442		
IP Address Assignme	nt Method (IPCP)		
Fixed IP :	⊙ No (Dynamic IP)   ○ Yes		
Fixed IP Address :			
Connection Detection	n		
Detect Interval :	10		
No-Reply Count :	2		
			Apply Reset Cancel

User Name Password	Assign a specific valid user name provided by local ISP. Assign a valid password provided by local ISP.
Authentication	Select <b>PAP</b> or <b>CHAP</b> protocol for widest compatibility. The default value is <b>PAP</b> . The password will be encrypted in CHAP but not in RAP.
MTU	It means the maximum transmission unit. Default value is 1442. Change it if you want.

<b>PPTP Local Address</b>	Assign a local IP address.
PPTP Subnet Mask	Assign a subnet mask value of IP address.
<b>PPTP Server Address</b>	Assign a remote IP address of PPTP server.
Fixed IP	Click <b>No</b> to use dynamic IP for PPPoE protocol; click <b>Yes</b> to use a fixed IP address. If you choose <b>Yes</b> , you have to type an IP address in the field of Fixed IP Address below.
Fixed IP Address	Type an IP address here if you click Yes as Fixed IP.
Detect Interval	Assign an interval time for detecting if the WAN connection is on or off.
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.
Apply	Click <b>Apply</b> to go back to the WAN Interface Configuration page. To apply all settings, click <b>Apply</b> 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 LAN

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

Network - LAN	
LAN IP/DHCP	
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 :	
Secondary DNS :	
Lease Time (Min) :	1440
Gateway IP(Optional) :	

In the VigorTalk ATA-24 adapter, some IP address settings could be configured for the LAN interface. The IP address/subnet mask is for private users or NAT users. The IP address of the

default gateway on other local PCs should be set as the VigorTalk ATA-24' server IP address. When the DSL connection between the DSL and the ISP has been established, each local PC can directly route to the Internet. The IP address/subnet mask can also be used to connect to other private users (PCs). On this page you will see the private IP address defined in RFC-1918. Usually we use the 192.168.1.0/24 subnet for the route.

IP Address	Type the IP address for LAN/DHCP.
Subnet Mask	Type the subnet mask for the LAN IP/DHCP.
Status	Click <b>Enable</b> the DHCP server; click <b>Disable</b> 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.

### 3.2.3 High Availability

The High Availability (HA) feature refers to the awareness of component failure and the availability of backup resources. The complexity of HA is determined by the availability needs and the tolerance of system interruptions. Systems, that provide nearly full-time availability, typically have redundant hardware and software that makes the system available despite failures.

The high availability is designed to avoid single points-of-failure. When failures occur, the failover process moves processing performed by the failed component (the "Master") to the backup component (the "Slave"). This process remains system-wide resources, recovers partial of failed transactions, and restores the system to normal within a matter of microseconds.

Take the following picture as an example. The left device is regarded as Master device, the right device is regarded as Slave device. When Master device is broken down, the Slave device could replace the Master role to take over all jobs as soon as possible. However, once the original Master is working again, the Slave would be changed to original role to stand by.



In the Network group, click the High availability option.

Network - LAN	- High Availability
High Availability:	O Disable 💿 Enable
Group Number:	1 (Range: 1~255)
Role:	Master 🗸
Virtrual IP :	192.168.1.3
	Apply Cancel

High Availability	Disables or enables this function. When the master device fails down, the slave device will take its work over.
Group Number	Assign a group number. The range is from 1 to 255. PCs on the same group (in LAN) can support for each other.
Role	Select a role for this device as Master or Slave.
Virtual IP	Assign an IP address as a virtual IP.

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 Static Route Setup

When you have several subnets in your LAN, sometimes a more effective and quicker way for connection is the **Static routes** function rather than other methods. You may simply set rules to forward data from one specified subnet to another specified subnet without the presence of RIP.

This function allows users to assign static routing information. In the **Advanced** group, choose **Static Route**. You will get the following page.

Network Interface	Gateway IP	Destination IP	Mask
۲			
0			
0			
0			
0			
)			
C			
0			
0			
0			

Network Interface	Displays the network interface (LAN, WAN1, 2, 3 or 4).
Gateway IP	Displays the gateway address of the static route.
Destination IP	Displays the destination IP of the static route.
Mask	Displays the subnet mask of this route.
Edit	Allows users to edit the selected static route settings.
Delete/Delete All	Removes one or all the selected static route settings.

The system allows users to set up to 10 static routes for the adapter.

### **Edit the Static Route**

To edit static route for certain item, select the radio button of the item and click **Edit** on the bottom of the page. The following web page will be displayed:

Advanced - St	atic Route - Edit
1	
Network Interface :	UAN1 💌
Gateway IP :	172.16.3.32
Destination IP :	202.66.88.99
Subnet Mask :	24 💌
	Apply Canc

Network Interface	Select a network interface as a destination to be sent. It includes LAN, and WAN1~WAN3.
Gateway IP	Assign an IP address of the gateway for the interface selected above.
Destination IP	Assign the IP address of the destination that data will be transferred to. Packets ready to destination will be sent out through the network interface chosen in this page.
Subnet Mask	Assign a value of subnet mask for destination IP address.

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

### **Delete the Static Route**

Select the radio button of the item that you want to delete and click **Delete** on the bottom of the page. The following web page will be displayed:

	Network Interface	Gateway IP	Destina	ation IP	Mask
	WAN1	172.16.3.32	202.66		/24
	0		Microsoft Internet Exp	olorer 👔	
	0		🔹 🕐 Are you sure y	rou want to delete this item?	?
	0		ОК	Cancel	
	0				
	0				
	0				
	0				
	0				
0	0				

Click **OK** to delete the entry in static route table.

Users can click **Delete All** to remove all entries in static route table.

### 3.3.2 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 supports ten port numbers to be blocked.

Index	Status	Port Number
1.	💿 Disable  🔘 Enable	
2.	💿 Disable  🔿 Enable	
3.	💿 Disable  🔿 Enable	
4.	⊙Disable ○Enable	
5.	⊙ Disable ○ Enable	
6.	💿 Disable  🔿 Enable	
7.	💿 Disable  🔿 Enable	
8.	💿 Disable  🔿 Enable	
9.	💿 Disable  🔿 Enable	
10.	💿 Disable  🔘 Enable	
		Apply Can
ex		The number of each entry.
tus		User can <b>Disable</b> or <b>Enable</b> the port block for the specifie port.

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

Click Apply to finish this setting.

### 3.3.3 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.

setting is 135.

	Domain Name	Server Provider	Server Type	Active	Status
		dyndns.org	dynamic	disable	Not Connected
		dyndns.org	dynamic	disable	Not Connected
		dyndns.org	dynamic	disable	Not Connected
		dyndns.org	dynamic	disable	Not Connected
		dyndns.org	dynamic	disable	Not Connected
		dyndns.org	dynamic	disable	Not Connected
		dyndns.org	dynamic	disable	Not Connected
		dyndns.org	dynamic	disable	Not Connected
		dyndns.org	dynamic	disable	Not Connected
)		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 :	O Disable 💿 Enable
Interface :	VAN1 🛩
Server Provider :	dyndns.org (www.dyndns.org)
Server Type :	dynamic 🗸
Domain Name :	abc. dyndns. org
Login Name :	draytek
Login Password :	•••••
Wild Card :	⊙ Disable ○ Enable
Backup MX:	⊙ Disable ○ Enable
Mail Extender :	dray@draytek.com
	Apply Cancel

Status	Click <b>Disable</b> to disable this function. Click <b>Enable</b> 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 supports 7 domain server providers as default.

	<pre>dyndns.org (www.dyndns.org)  dyndns.org (www.dyndns.org) no-ip.com (www.no-ip.com) DtDNS (www.dtdns.com) ChangeIP.com (www.changeip.com) dynamic-nameserver (www.dynamic-nameserver.com) huagai.net (www.ddns.cn) 3322 (www.3322.org)</pre>	
Server Type	Select <b>Static</b> , <b>Dynamic</b> or <b>Custom</b> type for this entry of DDNS settings.	3
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. T make things like www.yourhost.dyndns.org work), click "Enable" to active this function.	0
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.

### 3.3.4 Port Mirroring

VigorTalk-ATA Series supports port mirroring function in WAN interfaces. Generally speaking, this function copies traffic from one or more specific ports to a target port. This mechanism helps manager track the network errors or abnormal packets transmission without interrupting the flow of data access the network. By the way, user can apply this function to monitor all traffics which user needs to check.

There are some advantages supported in this feature. Firstly, it is more economical without other detecting equipments to be set up. Secondly, it may be able to view traffic on one or more ports within a VLAN at the same time. Thirdly, it can transfer all data traffics to be mirrored to one analyzer connect to the mirroring port. Last, it is more convenient and easy to configure in user's interface.

In the Advanced group, click the Port Mirroring option. You will see the following page.

Advanced - Po	rt Mirroring	
ODisable 💿 Enable		
Mirroring Port :	WAN 1 🛩	
Mirrored Port(s) :	CPU	
	IAN	
	✓ WAN 1	
	WAN 2	
	WAN 3	
		Apply Cancel

Enable/Disable	Click <b>Disable</b> to disable this function. Click <b>Enable</b> to activate this function.				
<b>Mirroring Port</b>	Select a port to view traffic sent from mirrored ports.				
Mirrored Port(s)	Click which ports are necessary to be mirrored.				

After finishing the settings, please click Apply.

# **3.4 Firewall Setup**

The firewall controls the allowance and denial of packets through the adapter. The **Firewall Setup** in the VigorTalk ATA-24 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.

Firewall - DoS							
DoS Defense : 🔿 Disable 💽 Enable							
Enable SYN flood defense :	Threshold: 300 Packets/sec Timeout: 10 sec						
Enable UDP flood defense :	Threshold: 300 Packets/sec Timeout: 10 sec						
Enable ICMP flood defense:	Threshold: 300 Packets/sec Timeout: 10 sec						
Enable Port Scan detection :	Threshold: 300 Packets/sec						
Block IP options	🗌 Block TCP flag scan						
Block Land	Block Tear Drop						
Block Smurf	Block Ping of Death						
Block trace route	Block ICMP fragment						
Block SYN fragment	Block Unknown Protocol						
Block Fraggle Attack							
	Apply Cancel						

DoS Defense	Enables or disables the DoS Defense function. The default value is <b>Disable</b> .			
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 <b>300</b> packets per second and <b>10</b> 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 <b>300</b> packets per second and <b>10</b> 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 <b>300</b> packets per second and <b>10</b> 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 <b>300</b> 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 <b>no flag scan</b> , <b>FIN without ACK scan</b> , <b>SYN FIN scan</b> , <b>Xmas scan</b> and <b>full Xmas scan</b> .				
Enable Tear Drop	activities include no flag scan, FIN without ACK scan, SYN				
Enable Tear Drop Enable Ping of Death	activities include <b>no flag scan</b> , <b>FIN without ACK scan</b> , <b>SYN</b> <b>FIN scan</b> , <b>Xmas scan</b> and <b>full Xmas scan</b> . 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				
-	<ul> <li>activities include no flag scan, FIN without ACK scan, SYN</li> <li>FIN scan, Xmas scan and full Xmas scan.</li> <li>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.</li> <li>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</li> </ul>				

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

sions.

# 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 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.

Vo	VoIP - Protocol								
Select Protocol : O SIP O MGCP O H248									
	SIP Server MGCP Configuration Configuratio			H248 ion Configuration					
SIP	Local Po	rt :	5060						
#	Active	Outbound Proxy	Proxy Name	Proxy Address	Proxy Port	Registrar Addr	Registrar Port	Expires (sec)	Domain
1.				0	5060	0	5060	3600	0
2.				0	5060	0	5060	3600	0
3.				0	5060	0	5060	3600	0
Exa	mple		iptel	iptel.org		iptel.org			iptel.org
Pro	xy	User-Ager	it Name						
1	. Dray	Tek ATA24-	1.1.0						
2. DrayTek ATA24-1.1.0									
3	3. DrayTek ATA24-1.1.0								
									Apply Cancel

### • 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.
<b>Registrar Address</b>	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 :	
	MGCP H248 figuration Configuration
CallAgent 1	
MGCP Call Agent Address :	192.168.100.100
MGCP Call Agent Port :	2727
CallAgent 2	
MGCP Call Agent Address :	0.0.0.0
MGCP Call Agent Port :	2727
General Setting	
MGCP Local Port :	2427
EndPoint Name Style :	
	○ aaln/#@
Logic ID Starting Number :	1 (Range:1 ~ 99999)
Wild-carded RSIP :	○ Each endpoint sends its own RSIP ④ Send only one wild RSIP
Range Wildcard RSIP :	🔿 Disable 💿 Enable
HeartBeat :	⊙ Disable ○ Enable
HeartBeat Period :	60 (Range:1 ~ 65535)
HeartBeat Retry :	3 (Range:1 ~ 300)
	Apply Cancel

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.					
	<b>Send only one wild RSIP</b> – 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 <b>Enable</b> to send out RSIP message (e.g., aaln/*@[172.16.3.5]). Click <b>Disable</b> to close such function.
HearBeat	Click <b>Enable</b> to check if MGCP server can work normally or not, otherwise click <b>Disable</b> .
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

Select Protocol :			1GCP 💽 H241	3
SIP Server Configuration		1GCP iguration	H248 Configura	
Call Agent				
H248 Call Agent Addres	в:	218.108.7.7	0	]
H248 Call Agent Port :		2944		
Message ID				
H248 MessageID Mode		💿 [IPAddre	ss]:Port ( (I	PAddres
H248 MessageID IP Mod	le :	💿 wan ip	Address 🔘 M	/anual
H248 MessageID Addre	BS :	210.21.21.1	1	]
H248 MessageID Port :		2944		
General Setting				
H248 Local Port :		2944		

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 <b>Message ID</b> field are used to be identified by the server supported with H248.
H248 MessageID IP Mode	If you choose <b>WAN IP Address</b> , the system will use WAN IP address configured in Network for such protocol; if you choose Manual IP Address, you have to type IP address manually in the field of <b>H248 MessageID Address</b> below.
H248 MessageID Address	Type WAN IP address used for the server supported with H248 when you choose <b>Manual IP address</b> as H248 MessageID Address.
H248 MessageID Port	The port number for MessageID.
H248 Local Port	The UDP port number in H248 local terminal.
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.

ŧ	User Name Proxy Server	VoIP IP Address	Ring Port	Ring Type Call Forwarding
۲	1001	WAN	1	All Ports
2 0	1002	WAN	2	All Ports
0	1003	WAN	3	All Ports
0	1004	WAN	4	All Ports
0	1005	WAN	5	All Ports
0	1006	WAN	6	All Ports
0	1007	WAN	7	All Ports
0	1008	WAN	8	All Ports

User Name	$1001 \sim 1032$ 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 interface for such account. Such interface is used to apply VoIP traffics. Display the ring port number when the specified SIP account
Call Forwarding	Blank - no call forwarding. Always - call forwarding for all of the calls.

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 A	Accounts - E	dit									
1											
O Disable 💿 E	nable										
Username:	1001										
Password:											
Display Name:	1001										
	thentication ID: 1001										
Proxy Server:	none	*									
Call without Registration:	<ul> <li>Disable</li> </ul>										
VoIP IP Address:	WAN 💌										
Call Forwardin	g										
💿 Disable											
O Callforwarding	i all calls										
O Callforwarding	i busy										
O Callforwarding	no answer after 3	rings (Ran	ge:1~10)								
SIP URL	0	(Exampl	le:8001@iptel.org)								
IncommingCall	Rings										
💿 Rings all	ports in the group	O Rings the	first available port	🔿 Rings by rou	und robin						
Ring Port Setti	ng										
P1	🔲 P2	P3	P4	P5	🔲 P6	P7	P8				
P9	🗌 P10	🗌 P11	P12	🔲 P13	P14	P15	P16				
🔲 P17	🔲 P18	P19	P20	🔲 P21	P22	P23	P24				
							Apply Cancel				

Disable/Enable	Click <b>Disable</b> to close this setting. Click <b>Enable</b> 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 <b>Enable</b> to invoke <b>Call</b> without register.
VoIP Address	The interface is used to apply VoIP traffics. There are two options: <b>WAN</b> and <b>LAN/VPN</b> . If LAN/VPN is selected, VoIP can be applied through a VPN tunnel to create a high security voice phone.

	WAN VAN LAN/VPN
Call Forwarding	There are four options for you to choose. <b>Disable-</b> It is to close call forwarding function.
	<b>Callforwarding busy-</b> It means the incoming calls will be forwarded into SIP URL only when the local system is busy.
	<b>Callforwarding no answer after-</b> It means if the incoming calls do not receive any response, they will be forwarded to the SIP URL by the time out.
	<b>SIP URL-</b> Type in the SIP URL (e.g., aaa@draytel.org or abc@iptel.org) as the site for call forwarded.
IncomingCall Rings	<b>Rings as all ports in the group-</b> All the ring ports selected for such account will ring when VigorTalk receives any incoming call.
IncomingCall Rings	such account will ring when VigorTalk receives any incoming
IncomingCall Rings	<ul><li>such account will ring when VigorTalk receives any incoming call.</li><li>Rings the first available port- The first ring port selected for such account will ring when VigorTalk receives any incoming</li></ul>

# 3.5.2 Port Settings

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

•0	1 <b>F - F</b>	TUIL SE	ettings							
#	Edit	Active	SIP Account	Call Waiting	Hotline	Mic/Spk Gain	FAX	Codec	DTMF	Port Locked
1	8	V	1 - 1001			0/0	T.38 Relay	G.729A	RFC2833	
2	3	V	2 - 1002			0/0	T.38 Relay	G.729A	RFC2833	
3	E.	V	3 - 1003			0/0	T.38 Relay	G.729A	RFC2833	
4	3	V	4 - 1004			0/0	T.38 Relay	G.729A	RFC2833	
5	E.	V	5 - 1005			0/0	T.38 Relay	G.729A	RFC2833	
6	3	V	6 - 1006			0/0	T.38 Relay	G.729A	RFC2833	
7	8	V	7 - 1007			0/0	T.38 Relay	G.729A	RFC2833	
8	3	V	8 - 1008			0/0	T.38 Relay	G.729A	RFC2833	
										123

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.

Port 1	
🔿 Disable 💿 Enable	
Default SIP Accounts:	1-1001
VoIP IP Address:	WAN 💌
Hotline	
Hotline Number to Internet:	
Codec	
Preferred Codec :	G.729A -8kbps
Single Codec :	
Codec Rate :	20 💌 (ms)
Codec VAD:	💿 Disable 🔿 Enable
CAS	
Microphone Gain:	0 (Range: -14 ~ 6)
Speaker Gain:	0 (Range: -14 ~ 6)
FAX	
FAX Mode:	T.38 Relay 🖌
DTMF	
DTMF Mode:	🔿 InBand 💿 OutBand(RFC2833) 🔿 SIP INFO Cisco 💌
DTMF Volume:	27 (Range: 0 ~ 31)
Call Waiting	
💿 Disable 🔘 Enable	
Port Unlock	
Manual Unlocked:	Unlock (Unlock the port when its call status is "Line Fault")

Port 1	Click <b>Enable</b> to activate this port or <b>Disable</b> to close this port. <b>Default SIP Accounts</b> – Choose one of the SIP account as the default setting. <b>VoIP IP Address -</b> The interface is used to apply VoIP traffics. There are two options: <b>WAN</b> and <b>LAN/VPN</b> . 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 supports five Codecs. The default setting is G.729A. You can choose another one as preferred Codec for outgoing calls. G.729A -8kbps G.711U(PCMU) -64kbps G.711A(PCMA) -64kbps G.729A -8kbps G.723.1 -6.3kbps G.726 -32kbps

**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.
	<b>Codec VAD- Enable</b> or <b>Disable</b> 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	<ul><li>Microphone Gain- The gain value while transmitting voice.</li><li>The default value is 0. The range is from -32 to 31.</li><li>Speaker Gain - The gain value while receiving voice. The default value is 0. The range is from -32 to 31.</li></ul>
FAX	<ul> <li>FAX Mode - The FAX function mode. There are three options:</li> <li><i>Transparent:</i> FAX will be transmitted via voice channel; no fax relay and no Codec change will be involved.</li> <li><i>T.38 Relay:</i> Using T.38 Fax Relay. This is the default value.</li> </ul>
DTMF	<ul> <li>DTMF Mode -</li> <li>InBand: Choose this one then the Vigor will send the DTMF tone as audio directly when you press the keypad on the phone.</li> <li>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.</li> <li>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.</li> <li>DTMF Volume – Determine the volume of DTMF voice signal. The more the number is set, the greater the sound is.</li> </ul>
Call Waiting	<b>Enable</b> – Activate the call waiting function. <b>Disable</b> – Close the call waiting function.
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 Series.

<del>4</del>	Speed Dial Phone Number	Speed Dial Destination	Memo
1	1001	1001@iptel.org	dial 1
2			
3			
4			
5			
Example	101	101@iptel.org	

**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 Advanced Speed Dial

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

≠ Prefix	Strip Length	Append	Destination	Memo
۲				
0				
3 🔘				
4 🔘				
5 🔘				
6 🔘				
7 🔘				
8 🔘				
9 🔿				
10 🔘				

Prefix	Displays the prefix number of the entry.
Strip Length	Displays the strip length of the entry.
Append	Displays the appended number of the entry.
Destination	Displays the IP address of the destination 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 - Advan	ced Speed Dial - Edit
1	
Prefix :	03
Strip Length:	0
Append :	886 ('p':delay 1.8sec)
Destination :	10.1.1.1
Memo :	number01
	Apply Cancel

Prefix	Assign a prefix for checking the phone number that users dial out. If the prefix of the outgoing call matches to the number set in this field, that outgoing call can apply the speed dial. For example, suppose that there are two outgoing calls with phone numbers of 03654321 and 04556890. In which, 03654321 is suitable for this speed dial rule.
Strip Length	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.
Append	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.
Destination	Assign an IP address for the destination which the SIP message would be sent to.
Memo	A description for 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.

Region : UK	~	Caller ID 7	Type: ETSI	~		
Tone Classfication	Low Frequency(Hz)	High Frequency(Hz)	TOn1 (10msec)	TOff1 (10msec)	TOn2 (10msec)	TOff2 (10msec)
Dial tone	350	440	500	0	0	0
Ringing tone	440	480	0	0	200	400
Busy tone	480	620	0	0	50	50
Congestion tone	480	620	0	0	25	25
Setting CPT Tone T	imer					
Dial Tone : 16	Busy Tone :	30 Howler To	ne : 60	Ringin	g Tone : 180	)
Special Dial 16	Call Waiting Tone :	30 Congestic	in 30	Reorde	er Tone : 30	

#### 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

phone.	
Finland	*
User Defined	
Australia	
Canada, US	
China	
Denmark	
Finland	
France	
Germany	
Hong Kong	
Japan	
Netherlands	
Norway	
Singapore	
Taiwan	
UK	

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 <b>User Defined</b> is selected in the <b>Region</b> field, users can select one of the supported values. If a country is selected, this field will display ID type value automatically. Caller ID Type : DTMF

DTMF	North Ame JAPAN ETSI	rica
	DTMF	

Setting CPT Tone TimerSet different timer for different tones to restrict the play time of<br/>tone. When the time is up, the tone broadcasting will be<br/>stopped.Dial Tone – A telephony signal which indicates that the status

**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.

**Ringing 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 Line Test

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



#### 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	

Return to line Test

### 3.5.7 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**, **VoIP FailOver, etc**.

TP Starting Port :	13456
'38 Mode :	🔿 Disable 💿 Enable
.38 Starting Port :	13456
.38 Redundancy number :	1 (Range: 0~4)
Dialing Completion Timeout :	4sec (Range: 1~60)
ine Polarity Reversal :	🗌 as Callee Answer 👘 as Callee On-HooK
/oIP FailOver :	Use POTS system
Echo Cancellation :	🔿 Disable 💿 Enable
Echo Cancellation Tail Length :	128 ms(Range: 8 ~ 128, should be multiple of 8)
Pulse Timing	
finimum pulse break time :	25 ms
flaximum pulse break time :	100 ms
finimum flash break time :	250 ms
Aaximum flash break time :	800 ms
/linimum pulse make time :	15 ms
flaximum pulse make time :	75 ms
/inimum pulse interdigit time :	250 ms
Required:	
1.Minimum pulse break time < Ma	xmum pulse break time < Minimum flash break time < Maxmum flash break time

<b>RTP Starting Port</b>	The starting port number for RTP protocol packet. The default setting is 13456.
T38 Mode	Click <b>Enable</b> to enable T.38 function. Click <b>Disable</b> 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	<ul><li>as Callee Answer - Check this box to generate line polarity reversal while the remote user picks up the phone call.</li><li>as Callee On-Hook - Check this box to generate line polarity reversal while the remote user hangs off the phone call.</li></ul>

VoIP FailOver	<b>Use POTS System</b> – When VoIP call is unavailable, the system will switch into PSTN phone automatically.
Echo Cancellation	Click <b>Enable</b> to cancel echo. Click <b>Disable</b> 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.
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.

Vol	P - Mis	cellaneous	- Ring Cad	lence					
#	Edit	Ton1	Toff1	Ton2	Toff2	Ton3	Toff3	Ton4	Toff4
1	3	1000 ms	4000 ms	0 ms	0 ms	0 ms	0 ms	0 ms	0 ms
2	3	800 ms	2000 ms	800 ms	4000 ms	0 ms	0 ms	0 ms	0 ms
3	<b>E</b>	800 ms	2000 ms	800 ms	4000 ms	0 ms	0 ms	0 ms	0 ms
4	3	800 ms	2000 ms	800 ms	4000 ms	0 ms	0 ms	0 ms	0 ms
5	3	800 ms	2000 ms	800 ms	4000 ms	0 ms	0 ms	0 ms	0 ms
6	3	800 ms	2000 ms	800 ms	4000 ms	0 ms	0 ms	0 ms	0 ms
7	3	800 ms	2000 ms	800 ms	4000 ms	0 ms	0 ms	0 ms	0 ms
8	<b>S</b>	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 :	1000	ms		
Toff1 :	4000	ms		
Ton2 :	0	ms		
Toff2 :	0	ms		
Ton3 :	0	ms		
Toff3 :	0	ms		
Ton4 :	0	ms		
Toff4 :	0	ms		
				Apply 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.

All Ports	Auto 💌						
🕽 One by One	Auto Modem						
Port 1:	Fax 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.

Port	Lock Status	Manual Control	Port	Lock Status	Manual Control	
1	Unlocked	Unlock Lock	13	Unlocked	Unlock Lock	
2	Unlocked	Unlock Lock	14	Unlocked	Unlock Lock	
3	Unlocked	Unlock Lock	15	Unlocked	Unlock Lock	
4	Unlocked	Unlock Lock	16	Unlocked	Unlock Lock	
5	Unlocked	Unlock Lock	17	Unlocked	Unlock Lock	
6	Unlocked	Unlock Lock	18	Unlocked	Unlock Lock	
7	Unlocked	Unlock Lock	19	Unlocked	Unlock Lock	
8	Unlocked	Unlock Lock	20	Unlocked	Unlock Lock	
9	Unlocked	Unlock Lock	21	Unlocked	Unlock Lock	
10	Unlocked	Unlock Lock	22	Unlocked	Unlock Lock	
11	Unlocked	Unlock Lock	23	Unlocked	Unlock Lock	
12	Unlocked	Unlock Lock	24	Unlocked	Unlock 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.

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:	8 10 mA	Port 8:	8 💌 mA	Port 14:	8 🔽 mA	Port 20:	8 💌 mA
Port 3:	12 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

All PortWhen you click this button and choose any number from the<br/>drop down list, all the configurations will be applied to all ring<br/>ports.8/10/12 - When the phone line current reaches<br/>8mA/10mA/12mA, the system will judge the phone is off-hook.One by OneWhen you click such item, you have to specify which ring port<br/>will be applied with the configuration set here. If you choose<br/>multiple ring ports, they will apply the configuration one by<br/>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.

🔿 All Ports	600 🔽 Ω						
🕑 One by One							
Port 1:	600 🔽 Ω	Port 7:	600 💌 Ω	Port 13:	600 💌 Ω	Port 19:	600 💌 Ω
Port 2:	600 900 Ω	Port 8:	600 🔽 Ω	Port 14:	600 💌 Ω	Port 20:	600 💌 Ω
Port 3:	<u>China</u> 600 <u>∨</u> Ω	Port 9:	600 🔽 Ω	Port 15:	600 💌 Ω	Port 21:	600 💌 Ω
Port 4:	600 💌 Ω	Port 10:	600 🔽 Ω	Port 16:	600 💌 Ω	Port 22:	600 💌 Ω
Port 5:	600 💌 Ω	Port 11:	600 💌 Ω	Port 17:	600 💌 Ω	Port 23:	600 💌 Ω
Port 6:	600 🔽 Ω	Port 12:	600 🗸 Ω	Port 18:	600 🗸 Ω	Port 24:	600 🗸 Ω

All PortWhen you click this button and choose any item from the drop<br/>down list, all the configurations will be applied to all ports.

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

#### Line PCM codec

There are two types, A-LAW and  $-\mu$  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 Ports	A-LAU 🗸						
💿 One by One							
Port 1:	A-LAW 🗸	Port 7:	A-LAW 🔽	Port 13:	A-LAW 🗸	Port 19:	A-LAW 🔽
Port 2:	A-LAW 💙	Port 8:	A-LAW 🔽	Port 14:	A-LAU 💙	Port 20:	A-LAW 🔽
Port 3:	A-LAW 💙	Port 9:	A-LAV 🔽	Port 15:	A-LAW 🗸	Port 21:	A-LAV 🔽
Port 4:	A-LAW 💙	Port 10:	A-LAV 🔽	Port 16:	A-LAU 💙	Port 22:	A-LAV 🔽
Port 5:	A-LAW 💙	Port 11:	A-LAV 🔽	Port 17:	A-LAW 🗸	Port 23:	A-LAW 🔽
Port 6:	A-LAV 🗸	Port 12:	A-LAW 🗸	Port 18:	A-LAV 🔽	Port 24:	A-LAV 🔽

All PortWhen you click this button and choose any item from the drop<br/>down list, all the configurations will be applied to all ports.600/900 – Available impedance value provided by the system.<br/>China - Such selection is available for the users in China.One by OneWhen you click such item, you have to specify which ring port

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

# 3.5.8 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.

Barring Cl	ass					
	Deny only	calls from o	deny list	~		
Match Met	thod					
	Name :	⊙ Disable Remind:	O Enable			
	IP/Domain :	⊙ Disable Remind :	O Enable			
Speed Dia	l 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. Deny only calls from deny list Allow all incoming calls Allow only calls from allow list Allow only calls from speed dial entries Deny only calls from deny list Deny all incoming calls
	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.
	<b>Deny only calls from deny list</b> – The calls listed on Deny List page will not be accepted by this adapter. And others calls are accepted.
	<b>Deny all incoming calls</b> – 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 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.

ŧ	Name	IP/Domain
	Tom	192.168.1.6
	John	iptel.org
xample	John	192.168.1.1 or iptel.org

Name	The name or number in the allow list.

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

#### **Deny List**

The VigorTalk ATA-24 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.

¥	Name	IP/Domain
1	James	172.16.3.221
2	Steven	arctel.com
3		
1		
5		
Example	John	192.168.1.1 or iptel.org

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.

#### 3.5.9 Statistics

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

🔃 Statistics 🔹 🕨	Call Statistics
	RTP Statistics
	🔧 RTP Threshold Setting
	Show Alert

#### **Call Statistics**

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

h Option:	No R	efresh 🔉	Refresh			
minutes	24 hours					
Successful Outg	joing Calls	Successful Incon	ning Calls	Failed Outgoing Calls	Failed Incoming Calls	Total Calls
	minutes		minutes 24 hours	minutes 24 hours	minutes 24 hours	minutes 24 hours

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 Re:	fre	sh 🔽	ļ
No Ref	res	sh	
Every	10	Seconds	
Every	20	Seconds	
Every	30	Seconds	

#### **RTP Statistics**

This page displays statistics for RTP.

Refresh	h Option:	No Refresh	✓ Refresh				
15 r	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							
10							
12							
13							
14							
15							
16							
17							
18							
19							
20							
21							
22							
23							

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 Re:	fre	sh	~
No Ref	res	sh	
Every	10	Seconds	
Every	20	Seconds	
Every	30	Seconds	

#### **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**.

Mode :	🔘 Disable 💿 Enable	
Round Trip Delay Low Threshold :	80	(ms)
Round Trip Delay High Threshold :	150	(ms)
Jitter Low Threshold :	3	(ms)
Jitter High Threshold :	10	(ms)
Packet Loss Ratio Low Threshold :	0	(0~100%)
Packet Loss Ratio High Threshold :	5	(0~100%)
RTCP Timeout :	10	(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.
<b>RTCP</b> Timeout	Set the value (default setting is 10) for RTP timeout setting.

### **Show Alert**

This page display information for alert message.

Refresh Option:	No Refresh	Refresh			
Port	Alert Count	Alert Type	Delay	Jitter	Lost
1		Unknow			
2		Unknow			
3		Unknow			
4		Unknow			
5		Unknow			
6		Unknow			
7		Unknow			
8		Unknow			
9		Unknow			
10		Unknow			
11		Unknow			
12		Unknow			
13		Unknow			
14		Unknow			
15		Unknow			
16		Unknow			
17		Unknow			
18		Unknow			
19		Unknow			
20		Unknow			
21		Unknow			
22		Unknow			
23		Unknow			

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 Re:	fre	sh	~
No Ref	res	sh	
Every	10	Seconds	
Every	20	Seconds	
Every	30	Seconds	

#### 3.5.10 Status

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



#### **Port Status**

This page displays the connection status for VoIP phone calls.

Refresh	Option:	No Refresh	Refresh	
#	Call Status		RTP Statistic	
1	Locked		PS=0, OS=0, PR=0, OR=0, PL=0, JI=0, LA=0	
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			

Call Status	Display the calling status, idle, far-end alerting, alerting, busy, dialing and connected.
RTP Statistics	The statistics for RTP. <b>PS</b> means packets sent; <b>OS</b> means octets sent; <b>PR</b> means packet received; <b>OR</b> means octets received; <b>PL</b> means packets lost, <b>LA</b> means average TX delay (unit is ms) and <b>JI</b> means inter arrival jitter estimates (unit is ms).

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 Re:	fre	sh	¥
No Ref	res	sh	
		Seconds	
Every	20	Seconds	
Every	30	Seconds	

#### **SIP Status**

This page displays the registration status for VoIP phone calls.

Refresh	Option:	No Refresh	✓ Refresh				
#	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.

No Refresh					
No Ref	res	sh			
Every	10	Seconds			
Every	20	Seconds			
Every	30	Seconds			

#### **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.

Refresh Op	otion: N	o Refresh	<ul> <li>Refresh</li> </ul>				
Device							
No Fault							
Port	Fault Status	Port	Fault Status	Port	Fault Status	Port	Fault Status
1	-	7	-	13	-	19	-
	-	8	-	14	-	20	-
2							
2	-	9	-	15	-	21	-
-	-	9 10	-	15 16	-	21	-
3							-

#### **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 Re:	fre	sh	~
No Ref	res	sh	
Every	10	Seconds	
Every	20	Seconds	
Every	30	Seconds	

#### 3.5.11 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.

Refrest	Refresh Option:			No Ret	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
	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=606, OR=12120, PL=0, JI=0, LA=2	G.729A 8kbps	20ms	Off	RFC283
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=606, OR=12120, PL=0, JI=0, LA=3	G.729A 8kbps	20ms	Off	RFC283
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	RFC283
l.	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=868, OR=17360, PL=0, JI=0, LA=3	G.729A 8kbps	20ms	Off	RFC283

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

]	No Rei	fre	sh	*
1	lo Ref	res	sh	
			Seconds	
I	Every	20	Seconds	
H	Every	30	Seconds	

The port number of VoIP.

Port Number

**Refresh Option** 

Start Time

**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.

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.

<b>Remote RTP Address</b>	The IP address of remote voice site.			
<b>Remote RTP Port</b>	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.			

#### 3.5.12 Configure Activate

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



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/LAN 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/LAN LED** is bright.



3. If not, there must be something wrong with the hardware connection. Simply back to **"2.1 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 stilled 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.

🕂 eth0 Properties 🔹 💽 🔀
General Authentication Advanced
Connect using:
ASUSTeK/Broadcom 440x 10/100 Ir
This connection uses the following items:
Client for Microsoft Networks     P. File and Printer Sharing for Microsoft Networks     Q. QoS Packet Scheduler     Thernet Protocol (TCP/IP)
Transmission Control Protocol/Internet Protocol. The default wide area network protocol that provides communication across diverse interconnected networks.
<ul> <li>✓ Show icon in notification area when connected</li> <li>✓ Notify me when this connection has limited or no connectivity</li> </ul>
OK Cancel

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

Internet Protocol (TCP)	/IP) Properties 🛛 💽 🔀
General Alternate Configu	ration
	ssigned automatically if your network supports you need to ask your network administrator for
Obtain an IP address	automatically
Use the following IP	address:
IP address:	
Sybnet mask:	
Default gateway:	
⊙ O <u>b</u> tain DNS server a	ddress automatically
Use the following DN	IS server addresses:
Preferred DNS server:	
Alternate DNS server:	
	Ad <u>v</u> anced
	OK Cancel

#### 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.

	Network	
how All Di	splays Sound Network Startup Disk	
	Location: Automatic	
	Show: Built-in Ethernet	
Config	gure IPv4: Using DHCP	
IP	Address: 192.168.1.10 Renew D	ICP Lease
Sub	net Mask: 255.255.255.0 DHCP Client ID: (If required (If	d)
DN	S Servers:	(Optional)
Search	Domains:	(Optional)
IPv6	Address: fe80:0000:0000:0000:020a:95ff:fe8d:72e4	
	Configure IPv6	(?)

# **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). The DOS command dialog will appear.

🕰 Command Prompt	- 🗆 ×
Microsoft Windows XP [Version 5.1.2600] (C) Copyright 1985-2001 Microsoft Corp.	<b>^</b>
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 ITL=255 Reply from 192.168.1.1: bytes=32 time<1ms ITL=255 Reply from 192.168.1.1: bytes=32 time<1ms ITL=255 Reply from 192.168.1.1: bytes=32 time<1ms ITL=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 = Oms, Maximum = Oms, Average = Oms	
D:\Documents and Settings\fae>_	

- 3. Type ping 192.168.1.1 and press [Enter]. It 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.
- Type ping 192.168.1.1 and press [Enter]. It the link is OK, the line of 64 bytes from 192.168.1.1: icmp\_seq=0 ttl=255 time=xxxx ms will appear.

00	Terminal — bash — 80x24	
Welcome to Darwin Vigor10:~ drayteks PING 192.168.1.1 64 bytes from 192 64 bytes from 192 64 bytes from 192 64 bytes from 192	an 3 02:24:18 on ttyp1 \$ ping 192.168.1.1 (192.168.1.1): 56 data bytes .168.1.1: icmp_seq=0 ttl=255 time=0.755 ms .168.1.1: icmp_seq=1 ttl=255 time=0.697 ms .168.1.1: icmp_seq=2 ttl=255 time=0.731 ms .168.1.1: icmp_seq=3 ttl=255 time=0.72 ms	R
5 packets transmit	ing statistics tted, 5 packets received, 0% packet loss g/max = 0.697/0.723/0.755 ms \$ ■	

# 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. **Network - WAN** Load Balance : O Disable ○ Enable ( ■ Auto Weight
   ) Backup: Edit IP Mode Defa # Active 2 WAN1 PPPoE ~ • R Γ WAN2 Not Set 0 R Γ 0 WAN3 Not Set

#### For PPPoE Mode

- 1. Check if **Username** and **Password** are entered with correct values that you **got from** your **ISP**.
- 2. Check if the setting of **Authentication** is correct or not. You may need to try both **PAP** and **CHAP**.
- 3. Check if Service Name (optional) is correct or not. It is required by some ISPs.



	PPoE/PPTP onfiguration		
User Name :	889966666@hinet.net	PPTP Local Address :	
Password :	•••••	PPTP Subnet Mask :	
Authentication :	PAP 💉	PPTP Server Address :	
Service Name :	hinet		
Connection Detec	tion		
Detect Interval :	10		
No-Reply Count:	2		
			Apply Reset Cancel

After finishing the settings, go to **System - Status** page and click **WAN Status**. You will get a correct web page of WAN settings.

Basic Status	LAN Status	WAN Status	
WAN1 :			
IP Address :	218.168.228.2	27	
MAC Address :	00:50:7f:28:80	D:e6	
Primary DNS :	168.95.1.1		
Secondary DNS :			
Gateway :	61.230.192.2	54	
RX Packets :	95		
TX Packets :	40		
Connection Status :	connected		
Up Time :	0 days 0 hour	s 4 minutes 45 seconds	
	Disconnec	t	

#### For Static Mode

 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.

	PPPoE/PPTP Configuration	
IP Address :	172.16.3.229	Host Name :
Subnet Mask :	255.255.255.0	Domain Name :
Default Gateway :	172.16.3.1	(Host Name and Domain Name are required for some ISPs.)
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.

Basic Status	LAN Status	WAN Status
WAN1 :		
IP Address :	220.130.52.2	21
MAC Address :	00:50:7f:28:8	0:e4
Primary DNS :	168.95.1.1	
Secondary DNS :		
Gateway :	220.130.52.2	209
RX Packets :	708	
TX Packets :	384	
Connection Status :	connected	
Up Time :	0 days 0 hou	rs 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.

	PPPoE/PPTP Configuration	
IP Address :	172.16.3.229	Host Name :
Subnet Mask :	255.255.255.0	Domain Name :
Default Gateway :	172.16.3.1	(Host Name and Domain Name are required for some ISPs.)
Primary DNS :		
Secondary DNS :		

- 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.

Basic Status	LAN Status	WAN Status
WAN1 :		
P Address :	172.16.100.10	)
MAC Address :	00:50:7f:28:80	):e5
Primary DNS :	172.16.100.1	
Secondary DNS :		
ateway:	172.16.100.1	
X Packets :	96	
X Packets :	100	
connection Status :	connected	
Jp Time :	0 days 0 hour	s 4 minutes 51 sec

#### For PPTP Mode

- 1. Check if the settings of Username and Password are correct or not.
- 2. Check if the setting of **Authentication** is correct or not. You may need to try both **PAP** and **CHAP**.
- 3. Check if the value of **PPTP Local Address**, **PPTP Subnet Mask**, and **PPTP Remote Address** are correct or not.

	E/PPTP guration		
User Name :	draytek	PPTP Local Address :	10.0.0.150
Password :	••••	PPTP Subnet Mask :	255.255.255.0
Authentication :	PAP 💌	PPTP Server Address :	10.0.0.137
Service Name :			

4. After finishing the settings, go to **System - Status** page and click **WAN Status**. You will get a correct web page of WAN settings.

Basic Status	LAN Status	WAN Status
WAN1 :		
IP Address :	61.230.208.2	02
MAC Address :	00:50:7f:28:8	D:e7
Primary DNS :	194.109.6.66	
Secondary DNS :	194.98.0.1	
Gateway :	61.230.208.2	45
RX Packets :	341	
TX Packets :	86	
Connection Status :	connected	
Up Time :	0 days 0 hour	rs 4 minutes 39 seconds
	Disconnec	t

# 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 20 seconds	
Reset to factory default	
	Apply

#### Hardware Reset

While the adapter is running (ACT LED blinking), press the **RST** button and hold for more than 5 seconds. When you see the ACT LED blinks rapidly, please release the button. 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 **<u>support@draytek.com</u>**.

# **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> advance
- Enter system diagnostics function ATA24> diag
- Enter firewall configuration function ATA24> firewall

- Enter network configuration function ATA24> network

- Enter system configuration function ATA24> system

- Enter voip configuration function ATA24> voip

# A.2.2 Other Commands

- Help

ATA24>?

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

# **A.3 Advance Commands**

#### A.3.1 General Commands

- Enter advance configuration function ATA24> advance

- Help in advance configuration function ATA24/ advance > ?

- Back to the root commands ATA24/ advance > ..

### A.3.2 Port Block Commands

- Display the status for port block setting ATA24/advance> block -s

- Enable port block setting ATA24/advance> <Index> <Enable> <Port number>

- Disable port block setting

ATA24/advance> <Index> <Disable>

<index></index>	Item number(1~10)
<disable enable=""></disable>	0: Disable
	1: Enable
<port number=""></port>	Available number 1 ~ 65535

#### A.3.3 Portmirror Commands

#### - Help

ATA24/advance> portmirror ?

- Display port mirror settings

ATA24/advance> portmirror -s

- Edit port mirror settings

ATA24/advance> <Enable> <Moirroring> <Mirror CPU> <Mirror LAN><Mirror WAN1> <Mirror WAN2> <Mirror WAN3>

<enable></enable>	0: Disable
	1: Enable
<moirroring></moirroring>	Moirroring Port
	1: WAN1
	2: WAN2
	3: WAN3
<mirror cpu=""></mirror>	0: Do not mirror,
<mirror lan=""></mirror>	1: Mirror
<mirror wan1=""></mirror>	
<mirror wan2=""></mirror>	
<mirror wan3=""></mirror>	

#### A.3.4 Staticroute Commands

- Help

ATA24/advance> staticroute ?

- Display static route settings
  - ATA24/advance> staticroute -s <Index>
- Edit static route settings

ATA24/advance> <Index> <Network Interface> <Destination IP> <Gateway IP> <Subnet Mask>

- Delete static route settings

staticroute -d <Index>

<index></index>	Item number(1~10)
<network interface=""></network>	0 : LAN
	1 : WAN1
	2 : WAN2
	3 : WAN3
<destination ip=""></destination>	IP address of the destination
<gateway ip=""></gateway>	IP address of the gateway
<subnet mask=""></subnet>	Available settings include: /24 ; /25 ; /26 ; /27 ; /28 ; /29 ;
	/30;/31;/32;/8;/9;/10;/11;/12;/13;/14;/15;/16;
	/17 ; /18 ; /19 ; /20 ; /21 ; /22 ; /23 ; /0

# **A.4 Diagnostics Commands**

#### A.4.1 General Commands

- Enter system diagnostics function ATA24> diag

- Help in the system diagnostics function ATA24/diag> ?

- Back to the root commands ATA24/diag> ..

#### A.4.2 Learning\_table Commands

- Help

ATA24/diag> learning\_table ?

- Learning table commands usage

#### ATA24/diag> Learning\_table

#### A.4.3 Netstat Commands

- Help

ATA24/diag> netstat ?

- Netstat commands usage ATA24/diag> netstat -h

- Netstat diagnostics utility ATA24/diag> netstat <cmd>

#### A.4.4 Nslookup Commands

- Help

ATA24/diag> nslookup ?

-Nslookup diagnostics utility

ATA24/diag> nslookup <IPorDomainName>

#### A.4.5 Ping Commands

- Help

ATA24/diag> ping ?

- Ping commands usage ATA24/diag> ping

- Ping diagnostics utility

#### ATA24/diag> ping <Source Interface> <Destination Address>

<source interface=""/>	0 : LAN
	1 : WAN1
	2 : WAN2
	3 : WAN3
<destination< th=""><th>Domain name or IP Address of destination</th></destination<>	Domain name or IP Address of destination
Address>	

#### A.4.6 Traceroute Commands

- Help

ATA24/diag> traceroute ?

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

<cmd> Octet string

# **A.5 Firewall Commands**

#### A.5.1 General Commands

- Enter firewall configuration function ATA24>firewall

- Help in the firewall function

#### ATA24/ firewall > ?

- Back to the root commands ATA24/ firewall > ..

#### A.5.2 DoS Commands

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

#### A.5.2.1 Icmpflood Command

- Help

ATA24/network>dos >icmpflood ?

- Icmpflood commands usage

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

#### ATA24/firewall/dos/icmpflood>threshold<Value> <Timeout>

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

#### A.5.2.2 Packetblock Command

#### - Help

#### ATA24/network>dos >packetblock ?

- Packetblock commands usage

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

<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 Fraggle attrack

#### A.5.2.3 Portscan Command

- Help

ATA24/network>dos >portscan ?

- Portscan commands usage

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

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

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

#### A.5.2.4 Synflood Command

- Help

ATA24/network>dos >synflood ?

- Portscan commands usage

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

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

<option></option>	0: disable SynFlood detection function	
	1: enable SynFlood detection function	
<value></value>	0-65535, default=300 packets/sec	
<timeout></timeout>	The value of time out	

#### A.5.2.5 Udpflood Command

- Help

ATA24/network>dos >udpflood ?

- Portscan commands usage

ATA24/firewall/dos/udpflood >enable <Option>

#### ATA24/firewall/dos/udpflood >threshold <Value>

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

#### A.5.2.6 Enable Command

- Help

ATA24/network>dos >enable ?

- Portscan commands usage

ATA24/firewall/dos >enable <Option>

<option></option>	0: disable DoS Function
_	1: enable DoS Function

# **A.6 Network Commands**

#### A.5.1 General Commands

- Enter network configuration function ATA24> network

- Help in the network diagnostics function ATA24/network> ?

- Back to the root commands ATA24/network> ...

#### A.5.2 LAN Commands

```
Help

ATA24/network>lan?

Set the dhcp server

ATA24/network/lan> dhcp

Set the IP NAT function

ATA24/network/lan> ip_nat

Set the IP route function

ATA24/network/lan> ip_route
```

#### A.5.2.1 DHCP Command

- Help

ATA24/network/lan>dhcp ?

- Display DHCP setting ATA24/network/lan>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>

<index></index>	1: LAN1	
	2: LAN2	
	3: LAN3	
<mode></mode>	0: Disable	
	1: Enable	
	2: Relay Agent	
<start ip=""></start>	IP address as starting point.	
<end ip=""></end>	IP address as ending point.	
<primary dns=""></primary>	IP address as primary DNS.	

IP address as secondary DNS.	
IP address as gateway.	
Unit is minute.	
1: WAN1	
2: WAN2	
3: WAN3	
IP address as DHCP server.	

#### A.5.2.2 IP\_Nat Command

- Help

ATA24/network/lan>ip\_nat ?

- Display nat setting

ATA24/network/lan>ip\_nat -s <Index>

- Edit IP\_NAT setting

ATA24/network/lan>ip\_nat <Index> <Address> <Netmask>

<index></index>	1: LAN1 2: LAN2 3: LAN3
<address></address>	IP address for NAT.
<netmask></netmask>	Subnet mask for NAT.

#### A.5.2.3 IP\_Route Command

- Help

ATA24/network/lan>ip\_route ?

- Display IP route setting

ATA24/network/lan>ip\_route -s <WAN Interface>

- Edit IP\_Route setting

ATA24/network/lan> ip\_route -disable <WAN Interface>

ATA24/network/lan> ip\_route -enable <WAN Interface> <Address> <Netmask> <LAN Interface>

<pre></pre>   <	1: WAN1	
	2: WAN2	
	3: WAN3	
<address></address>	IP address for IP route.	
<netmask></netmask>	Subnet mask for IP route.	
<lan interface=""></lan>	1: LAN1 2: LAN2 3: LAN3	

#### A.5.3 WAN Commands

- Help

ATA24/network/wan?

#### A.5.3.1 Load Balance for WAN Command

- Help

ATA24/network/wan>advance> loadbalance ?

#### - Display the setting

ATA24/network/wan>advance> loadbalance -s

- Eidt the setting

ATA24/network/wan>advance> loadbalance <status> <autoweight>

<status></status>	0: Disable 1: Enable
<autoweight></autoweight>	0: Disable
	1: Enable

#### A.5.3.2 Backup Configuration Command

- Help

ATA24/network/wan>advance> backup?

- Display the setting

ATA24/network/wan>advance> backup -s

- Eidt the setting

ATA24/network/wan>advance> backup <status>

<status></status>	0: Disable
	1: Enable

#### A.5.3.3 Weight Configuration Command

- Help

ATA24/network/wan>advance> weight?

- Display the setting

ATA24/network/wan>advance> weight -s

- Edit weight setting

ATA24/network/wan>advance> weight <WAN1> <WAN2> <WAN3>

<wan1></wan1>	1:10%
<wan2></wan2>	2:20%
<wan3></wan3>	3: 30%
	4:40%
	5: 50%
	6: 60%
	7: 70%
	8:80%
	9: 90%

#### A.5.3.4 Set WAN to Active Command

- Help

ATA24/network/wan>active ?

- Edit WAN setting

ATA24/network/wan>active <index> <status> <default route> ATA24/network/wan>active <index> <status> <default route> <loadbalance><backupmaster> <backupslave>

<index></index>	1: WAN1
	2: WAN2
	3: WAN3
<status></status>	0: not active
	1: active
<default route=""></default>	0: not default
	1: default
<loadbalance></loadbalance>	0: not join loadbalance
	1: join loadbalance
<backupmaster></backupmaster>	0: not backupmaster
	1: backupmaster
<backupslave></backupslave>	0: not backupslave
	1: backupslave

#### A.5.3.5 Set WAN to DHCP Mode Command

- Help

ATA24/network/wan>dhcp ?

- Display current setting

ATA24/network/wan>dhcp -s <index>

- Edit WAN setting

#### ATA24/network/wan>dhcp <index> ATA24/network/wan>dhcp <index> <hostname> <domainname>

<index></index>	1: WAN1
	2: WAN2
	3: WAN3
< hostname >	Name of the host.
< domainname >	Name of the domain

#### A.5.3.6 Configure MAC Address Command

- Help

ATA24/network/wan>mac ?

- Display current setting

ATA24/network/wan>mac -s <index>

- Edit WAN setting

ATA24/network/wan>mac <index> <Use Default>

#### ATA24/network/wan>mac <index> <User Define> <Mac Address>

<index></index>	1: WAN1
	2: WAN2
	3: WAN3
<use default=""></use>	0: use default setting
<user define=""></user>	1: user defined setting
<mac address=""></mac>	MAC address for user defined configuration

#### A.5.3.7 PPPoE/PPTP Connection Detection Command

- Help

#### ATA24/network/wan>ppp\_detect ?

#### - Display current setting

#### ATA24/network/wan> ppp\_detect -s <index>

#### - Set condition for detection

ATA24/network/wan> ppp\_detect <index> <detect interval> <No-Reply Count>

<index></index>	1: WAN1
	2: WAN2
	3: WAN3
<detect interval=""></detect>	Assign a number as interval time for detecting.
<no-reply count=""></no-reply>	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.

#### A.5.3.8 Set WAN to PPPoE Mode Command

- Help
  - ATA24/network/wan>pppoe ?
- Display current setting

ATA24/network/wan> pppoe -s <index>

- Edit WAN setting

# ATA24/network/wan> pppoe <index> <Username> <Password> <Authentication Mode> <Service Name>

<index></index>	1: WAN1
	2: WAN2
	3: WAN3
<username></username>	Name (user account) assigned by ISP.
<password></password>	Password assigned by ISP.
<authentication< th=""><th>0:PAP</th></authentication<>	0:PAP
Mode>	1:CHAP
<service name=""></service>	Name (service) assigned by ISP.

#### A.5.3.9 Set WAN to PPTP Mode Command

#### - Help

ATA24/network/wan>pptp ?

- Display current setting

ATA24/network/wan> pptp -s <index>

- Edit WAN setting

ATA24/network/wan> pptp <index> <Username> <Password> <Authenticate Mode> <Local IP><Local Netmask> <Server IP>

<index></index>	1: WAN1
	2: WAN2
	3: WAN3
<username></username>	Name (user account) assigned by ISP.

<password></password>	Password assigned by ISP.	
<authentication< th=""><th>0:PAP</th></authentication<>	0:PAP	
Mode>	1:CHAP	
<local ip=""></local>	IP address for local host.	
<local netmask=""></local>	Netmask address for local host.	
<server ip=""></server>	IP address for the PPTP server.	

#### A.5.3.10 Configure UP/Downstream Rate Command

- Help

ATA24/network/wan>rate ?

- Display current setting

ATA24/network/wan> rate -s <index>

- Edit WAN setting

ATA24/network/wan> rate <index> <Downstream> <Upstream>

<index></index>	1: WAN1
	2: WAN2
	3: WAN3
<downstream></downstream>	0: using default setting (102400)
	Type any number to set downstream rate.
<upstream></upstream>	0: using default setting(102400)
	Type any number to set upstream rate.

#### A.5.3.11 Show WAN Configuration Command

- Help

ATA24/network/wan>show ?

- Display all WAN interfaces settings ATA24/network/wan> show

- Display specified WAN interface settings ATA24/network/wan>show <index>

<index></index>	1: WAN1
	2: WAN2
	3: WAN3
<index></index>	1: WAN1
	2: WAN2
	3: WAN3

#### A.5.3.12 Configure WAN Speed Command

- Help

ATA24/network/wan>speed ?

- Display current setting

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

- Edit WAN setting

```
ATA24/network/wan>speed <index> <Speed & Duplex>
```

<index></index>	1: WAN1	
	2: WAN2	
	3: WAN3	

<speed &="" duplex=""></speed>	1:Auto Negotiation 2:100M / Full Duplex 3:100M / Half Duplex 4:10M / Full Duplex 5:10M / Half Duplex
	2:100M / Full Duplex
	3:100M / Half Duplex
	4:10M / Full Duplex
	5:10M / Half Duplex

#### A.5.3.13 Set WAN to Static Mode Command

- Help

ATA24/network/wan>static ?

- Display current setting

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

- Edit WAN setting
  - ATA24/network/wan> static <index> <IP> <Netmask> <Gateway> <Primary DNS> <Secondary DNS>

<index></index>	1: WAN1
	2: WAN2
	3: WAN3
<ip></ip>	Private IP address for WAN.
<netmask></netmask>	Subnet mask for WAN.
<gateway></gateway>	Private IP address for gateway.
<primary dns=""></primary>	Pprivate IP address as primary DNS.
<secondary dns=""></secondary>	Private IP address as secondary DNS.

#### A.5.3.14 Static Connection Detection Command

- Help

ATA24/network/wan>static\_detect ?

- Display current setting

ATA24/network/wan> static\_detect -s <index>

- Set condition for detection, sending ARP to Gateway

ATA24/network/wan> static\_detect <index> 0 <detect interval> <No-Reply Count>

- Set condition for detection, sending PING

ATA24/network/wan> static\_detect <index> 1 <detect interval> <No-Reply Count> <detect destination>

- Set condition for detection, sending HTTP

ATA24/network/wan> static\_detect <index> 2 <detect interval> <No-Reply Count> <detect destination>

<index></index>	1: WAN1
	2: WAN2
	3: WAN3
<detect interval=""></detect>	Assign a number as interval time for detecting.
<no-reply count=""></no-reply>	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=""></detect>	Private IP address or domain name
# **A.6 System Commands**

# A.6.1 General Commands

- Enter system configuration function ATA24> system

- Help in the system configuration function ATA24/system> ?

- Back to the root commands ATA24/system> ..

### A.6.2 View ARP Cache Table Command

- Help

ATA24/system/DiagnosticTools>arpcachetable ?

- Display the setting

ATA24/system/DiagnosticTools> arp cache table

### A.6.3 View DHCP Assignment Command

- Help

ATA24/system/DiagnosticTools> dhcpassignmenttable ?

- Display the setting

ATA24/system/DiagnosticTools> dhcp assignment table

#### A.6.4 View Routing Table Command

- Help

ATA24/system/DiagnosticTools> routingtable ?

- Display the setting

ATA24/system/DiagnosticTools> routing table

## A.6.5 Administrator Control Commands

- Help

ATA24/system> administrator ?

- Edit password for administrator

ATA24/system>administrator<old password> <new password>

<verify password>

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

## A.6.6 Auto Logout Commands

- Help

ATA24/system > auto\_logout ?

- Display the setting

ATA24/system > auto\_logout -s

- Edit the max-cli-session number

ATA24/system > auto\_logout -n <MaxSess>

#### - Kill the #'s log-session

ATA24/system > auto\_logout -d <SessNum>

- Edit the maximum idle time of auto logout

#### ATA24/system > auto\_logout -m <MaxIdleTime>

- Enable/Disable the auto logout

ATA24/system > auto\_logout <Active>

<maxsess></maxsess>	Integer(1 to15)
<sessnum></sessnum>	Integer(1 to MaxSess)
<maxidletime></maxidletime>	Seconds, Integer(10 to 86400)
<active></active>	0: Disable 1: Enable

# A.6.7 Config Commands

- Help

ATA24/system> config ?

- Display the setting ATA24/system> config -s
- Execute the backup action

ATA24/system> config backup <fname> <servIP>

- Execute the restore action

ATA24/system> config restore <fname> <servIP>

<fname></fname>	Octets string maximum length is 64.
<servip></servip>	IP address for the IVD

## A.6.8 Manage Port Commands

#### - Help

ATA24/system> manage\_port ?

- Display the setting

ATA24/system> manage\_port -s

- Manage port from WAN interface

ATA24/system> manage\_port -m <Use Default Port or Not><Manage from WAN>

- Reboot the system to apply the changes ATA24/system> manage port -r

- Enable HTTP/Telnet function

```
ATA24/system> manage_port -e <HTTP Enable> <TELNET Enable>
```

- Change port number for HTTP/Telnet function ATA24/system> manage\_port -p <Http> <Telnet>

- Set IP address for the connection through WAN interface ATA24/system> manage\_port -i <index> <IP Start> <IP End>

<b><use b="" default="" or<="" port=""></use></b>	0 : Default
Not>	1 : User Define
<http></http>	default: 80
<telnet></telnet>	default: 23
<manage from<="" th=""><th>0 : Disable all from Wan;</th></manage>	0 : Disable all from Wan;

WAN>	1 : Enable all from Wan; 2 : Enable only defined Wan IP;
<ip start=""></ip>	Starting point
<ip end=""></ip>	Ending point.

## A.6.9 Reboot Commands

- Help

ATA24/system> reboot ?

- Reboot the system ATA24/system> reboot

 Reboot the system with keeping some important configuration ATA24/system> reboot keep

 Reboot the system with factory default configuration ATA24/system> reboot default

 Reboot the IVD VoIP board only ATA24/system> reboot voip ATA24/system> reboot dsl

## A.6.10 Show Status Command

- Help

ATA24/system> status ?

- Display the system status ATA24/system> status

#### A.6.11 Syslogd Commands

- Help

ATA24/system> syslogd ?

- Display the syslog setting ATA24/system> syslogd -s

- Set IP address and port number for Syslog server

ATA24/system>syslogd <Active> <RIP> <RPort> <Facility> <Severity>

<active></active>	0: Disable
	1: Enable
<rip></rip>	Type IP address for LAN
<rport></rport>	Integer(1 to 65535)
<facility></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></severity>	0: Emergency(default settting)
	1: Alert
	2: Critical
	3: Error
	4: Warning

5: Notice (including SIP)

- 6: Informational
- 7: Debug

# A.6.13 Upgrade Commands

#### - Help

ATA24/system> upgrade ?

- Display the setting

ATA24/system> upgrade -s

- Execute the firmware upgrade

### ATA24/system> upgrade <File Name> <Server IP>

<file name=""></file>	Octets string maximum length is 64.
<server ip=""></server>	Type IP address for the IVD.

# A.7 Voip Commands

### A.7.1 General Commands

- Enter voip configuration function ATA24> voip

- Help in the voip diagnostics function ATA24/voip> ?

- Back to the root commands ATA24/voip>..

# A.7.2 H248 Commands

- Help

#### ATA24/voip>h248?

- Display H248 call agent setting ATA24/voip/h248 > callagent -s

- Edit the H248 call agent setting ATA24/voip/h248>callagent <IPAddress> <Port>

- Display digit map default short/long timer setting ATA24/voip/h248 >dmTimer -s

- Edit the digit map timer setting ATA24/voip/h248>dmTimer <Termination> <Timer> <Sec>
- Display local listening port number for H248 ATA24/voip/h248 >localport -s
- Edit the local listening port setting ATA24/voip/h248>localport <Port>

- Display message ID ATA24/voip/h248 >mid -s

- Edit message ID

ATA24/voip/h248>mid -m <Mode> ATA24/voip/h248>mid -i <IP Mode> ATA24/voip/h248>mid <IPAddress> ATA24/voip/h248>mid <IPAddress> <Port>

- Display termination ID ATA24/voip/h248 >termid -s

- Edit termination ID

ATA24/voip/h248>termId -a <Prefix> <StartNum> ATA24/voip/h248>termId <Termination> <ID>

<ipaddress></ipaddress>	Domain name or IP Address
<port></port>	1 to 65535
<termination></termination>	1 to 24
<timer></timer>	0: short timer
	1: long timer
<sec></sec>	1 to 99 (sec)
<mode></mode>	0: [IPAddress]:Port
	1: [IPAddress]
<ip mode=""></ip>	0: WAN IPAddress

1: Manual IPAddress
ID Name prefix
Beginning of ID Name Number
1 to 24
Identification name

## A.7.3 Linetest Commands

### - Help

ATA24/voip>linetest ?

- Execute voip line card test

ATA24/voip/linetest > line\_card\_test <Line> <TestItem>

- Execute voip metallic loop test

ATA24/voip/linetest > metallic\_loop\_test <Line>

- Execute voip user phone test

ATA24/voip/linetest >user\_phone\_test <Line> <TestItem>

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

# A.7.4 MGCP Commands

- Help

ATA24/voip/mgcp> callagent ?

- Display the call agent setting

ATA24/voip/mgcp> callagent -s ATA24/voip/mgcp> callagent2 -s

- Edit the IP address and port number for call agent ATA24/voip/mgcp> callagent <IPAddress> <Port>

#### ATA24/voip/mgcp> callagent2 <IPAddress> <Port>

- Display the setting of End Point Name ID Style ATA24/voip/mgcp> epidstyle -s

- Edit the style mode for end point

ATA24/voip/mgcp> epidstyle -m<Mode>

- Edit the logic ID for end point ATA24/voip/mgcp> epidstyle -l <LogicID>
- Edit the domain name for end point ATA24/voip/mgcp> epidstyle -d <DomainName>
- Display the MGCP heartbeat setting

# ATA24/voip/mgcp> heartbeat -s

<ul> <li>Edit the dual_homing action ATA24/voip/mgcp&gt; heartbeat <active></active></li> </ul>
<ul> <li>Edit the period of heartbeat for dual_homing</li> <li>ATA24/voip/mgcp&gt; heartbeat -t <sec></sec></li> </ul>
- Edit the retry times of dual_homing ATA24/voip/mgcp> heartbeat -r <times></times>
- Display local port setting ATA24/voip/mgcp> localport -s
<ul> <li>Edit the local port number for MGCP protocol ATA24/voip/mgcp&gt; localport <port></port></li> </ul>
- Display the port lock setting ATA24/voip/mgcp> portlock –s
ATA24/voip/mgcp> portlock -s <port></port>
<ul> <li>Edit the port lock/unlock</li> <li>ATA24/voip/mgcp&gt; portlock <port> <lock></lock></port></li> </ul>
- Display the setting ATA24/voip/mgcp> rsip -s
- Set the RSIP action ATA24/voip/mgcp> rsip <active></active>
<ul> <li>Display the setting pf sending RSIP with wildcarded endpoint ID ATA24/voip/mgcp&gt; rsip -s</li> </ul>
<ul> <li>Edit the RSIP action ATA24/voip/mgcp&gt; wildrsip <wildcard> <range></range></wildcard></li> </ul>
<ipaddress> Assign an IP address of Call Agen</ipaddress>

<ipaddress></ipaddress>	Assign an IP address of Call Agent server in
	MGCP (Default is 192.168.100.100)
<port></port>	Assign a UDP port number to Call Agent server.
	1 to 65535 (Default is 2727)
<mode></mode>	There are four options for users to select. (Default
	is 0)
	0. <u>aaln/#@[ip_addr]</u>
	ex: aaln/1@[1.1.1.1]
	1. <u>mac_addr/#@[ip_addr]</u>
	ex: 000504030201/1@[1.1.1.1]
	2. aaln/#@mac_addr
	ex: aaln/1@000504030201
	3. aaln/#@domain_name
	ex: <u>aaln/1@callagent.com</u>
<logicid></logicid>	Starting number for logic ID.
<domainname></domainname>	Name of the domain
<active></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></sec>	Integer(1 to 65535 default=60)

<times></times>	Integer(1 to 300 default=1)
<port></port>	1 to 24
(for port lock/unlock)	
<lock></lock>	0: unlocked (default)
	1: locked
<wildcard></wildcard>	1: Enable wildcard(*) RSIP(Default)
	0: Disable wildcard(*) RSIP
<range></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/voip>misc ?
- Display the dialing completion timeout ATA24/voip/misc> dialing\_timeout -s
- Set the dialing completion timeout ATA24/voip/misc> dialing\_timeout <value>
- Display echo cancellation configuration ATA24/voip/misc> echo\_cancellation -s
- Enable echo cancellation configuration
   ATA24/voip/misc> echo\_cancellation <enable>

ATA24/voip/misc> echo\_cancellation <enable> <tailLength>

- Display VoIP failover configuration ATA24/voip/misc> failover -s
- Enable/disable VoIP failover configuration when it failed from network to gateway ATA24/voip/misc> failover –n <Mode>
- Enable/disable VoIP failover configuration to use POTS system forcefully ATA24/voip/misc> failover –f <Mode>
- Display gain control setting ATA24/voip/misc>gain -s
- Set gain control setting ATA24/voip/misc>gain <Device port> <Speaker Gain> <Microphone Gain>
- Display line impedance parameter ATA24/voip/misc> lineimpedance -s
- Set same value for each line ATA24/voip/misc> lineimpedance <Country>

ATA24/voip/misc>lineImpedance <line> <Country>

- Display line PCM codec ATA24/voip/misc> linepcmcodec -s
- Set same value for each line ATA24/voip/misc> linepcmcodec <codec>

ATA24/voip/misc> linepcmcodec <line> c

- Display metering parameter

ATA24/voip/misc> metering -s

- Set metering parameter

ATA24/voip/misc> metering

ATA24/voip/misc> metering -r <Reversal as Callee off-hook> <Reversal as Callee on-hook>

- Display NAT traversal setting ATA24/voip/misc> nat -s
- Set NAT traversal setting

ATA24/voip/misc>nat <Disable Mode>

ATA24/voip/misc>nat <Manual Mode> <NatIpAddr>

ATA24/voip/misc>nat <Auto Mode> <Type> <LocalPort> <ServerIP> <ServerPort>

ATA24/voip/misc>nat -sym <sym\_rtp\_t38>

- Display Line offhook detect current value ATA24/voip/misc>offhookdetect -s
- Set Line offhook detection ATA24/voip/misc> offhookdetect <Current>

ATA24/voip/misc> offhookdetect <line> <Current>

- Display pulse timing configuration ATA24/voip/misc> pulsetime -s
- Set pulse timing

ATA24/voip/misc> pulsetime <breakMin> <breakMax> <flashMin> <flashMax> <makeMin> <makeMax> <interdigitmin>

- Display ring cadence and frequency setting ATA24/voip/misc> ring -s
- Set ring cadence and frequency (same value for each line) ATA24/voip/misc> ring -f <Frequency>
- Set ring frequency ATA24/voip/misc>ring -f <line> <Frequency>
- Set ring cadence

ATA24/voip/misc>ring -c <Index> <Ton1> <Toff1> <Ton2> <Toff2> <Ton3> <Toff3> <Ton4> <Toff4>

- Display the port number for sending/receiving RTP packets ATA24/voip/misc> rtp\_port -s
- Set the port number for sending/receiving RTP packets ATA24/voip/misc> rtp\_port <Port number>
- Display T.38 Fax Relay Configuration ATA24/voip/misc> t38 -s
- Set T.38 Fax Relay ATA24/voip/misc> t38 <Mode>

#### ATA24/voip/misc t38 <Mode> <Port> <Redundancy>

- Display Voice Band Data (VBD) Configuration ATA24/voip/misc> vbd -s

- Set Voice Band Data (VBD) (same value for each line)

## ATA24/voip/misc> vbd <VBD>

## - Set Voice Band Data (VBD)

ATA24/voip/misc> vbd <port><vbd></vbd></port>	>
---	---

<value></value>	Range: 1~60 (second)
<enable></enable>	0: disable
	1: enable
<taillength></taillength>	Network Echo Canceller Tail Length (ms)
C	Range: $8 \sim 128$ , should be multiple of 8
<mode></mode>	0: disable
	1: enable
<device port=""></device>	Device port number
<speaker gain=""></speaker>	Assign the gain value while receiving voice,
-pound our	default value is 0.
	The range is from -14 to 6.
<microphone gain=""></microphone>	Assign the gain value while transmitting voice,
interophone Guin	default value is 0.
	The range is from -14 to 6. (Default is 0)
<li>line&gt;</li>	Device line number (from 1 to 24)
<country></country>	0: 600 Ohm (default)
······································	1: 900 Ohm
	2: China
<codec></codec>	0: Mu-LAW (default)
	1: A-LAW
<reversal as="" callee="" off-hook=""></reversal>	0: Disable (default)
-Keversar as Canee on-nook-	1: Enable
<reversal as="" callee="" on-hook=""></reversal>	0: Disable (default)
-Keversar as Canee on-nook-	1: Enable
<disable mode=""></disable>	0 : Disable NAT traversal (DEFAULT)
<manual mode=""></manual>	1 : Manually input NAT IP address
<auto mode=""></auto>	2 : Auto discover NAT IP address
<natlpaddr></natlpaddr>	Type IP address for manual mode.
	0 : Semi-auto, need to configure NAT
<type></type>	1 : Full-auto, no need to configure NAT
<localport></localport>	
<pre><localfort></localfort></pre>	Local listening port number for STUN client The IP address of STUN server
<pre><serverport></serverport></pre>	
	The port number of STUN server
<sym_rtp_t38></sym_rtp_t38>	0 : Disable symmetric RTP and T.38
< <u></u>	1 : Enable symmetric RTP and T.38
<current></current>	8: 8 mA (default)
	10: 10 mA
	12: 12 mA
change le Mines	15: 15 mA
<pre><breakmin> </breakmin></pre>	Minimum pulse break time (ms)
<pre><breakmax></breakmax></pre>	Maximum pulse break time (ms)
<pre><flashmin></flashmin></pre>	Minimum flash break time (ms)
<flashmax></flashmax>	Maximum flash break time (ms)
<makemin></makemin>	Minimum pulse make time (ms)
<makemax></makemax>	Maximum pulse make time (ms)
<interdigitmin></interdigitmin>	Minimum pulse inter digit time (ms)
<freqnency></freqnency>	Ring frequency
	20: 20 HZ (default)
	25: 25 HZ

<index></index>	Pattern Index, Index Value: 1-8	
<ton1></ton1>	Ton1 of cadence, unit: (ms)	
<toff1></toff1>	Toff1 of cadence, unit: (ms)	
<ton2></ton2>	Ton2 of cadence, unit: (ms)	
<toff2></toff2>	Toff2 of cadence, unit: (ms)	
<ton3></ton3>	Ton3 of cadence, unit: (ms)	
<toff3></toff3>	Toff3 of cadence, unit: (ms)	
<ton4></ton4>	Ton4 of cadence, unit: (ms)	
<toff4></toff4>	Toff4 of cadence, unit: (ms)	
<port number=""></port>	1 to 65535	
<mode></mode>	0: Disable	
	1: Enable	
<port></port>	T.38 Starting Port, 1 to 65535 (default:13456)	
<redundancy></redundancy>	T.38 Redundancy Number, 0 to 4 (default:1)	
<port></port>	device port number	
<vbd></vbd>	0: Auto Detection	
	1: Modem	
	2: Fax	

**Note:** "Auto Discovery NAT IP Address" option is used when IVD is behind a NAT adapter, NAT uses dynamic WAN IP address like as DHCP or PPPoE client. There must be having a STUN server in Internet. IVD 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/voip/sip> ?
- Enter incallbarring configuration function ATA24/voip/sip> incallbarring
- Display allow list of incoming calls (for SIP) ATA24/voip/misc> allow -s

ATA24/voip/misc>allow -s <Index>

- Edit allow list of incoming calls (for SIP) ATA24/voip/sip> allow -e <Index> <Name> <IP/Domain>
- Delete allow list of incoming calls (for SIP) ATA24/voip/sip> allow -d <Index>

ATA24/voip/sip> allow -d

- Display deny list of incoming calls (for SIP) ATA24/voip/misc> deny -s

#### ATA24/voip/misc> deny -s <Index>

- Edit deny list of incoming calls (for SIP) ATA24/voip/sip> deny -e <Index> <Name> <IP/Domain>

- Delete deny list of incoming calls (for SIP) ATA24/voip/sip> deny -d <index></index>
ATA24/voip/sip> deny -d
- Display current settings for incoming call barring (for SIP) ATA24/voip/misc> set -s
<ul> <li>Edit deny list of incoming calls (for SIP) ATA24/voip/sip&gt; set <class> <matchname> <matchip> <speeddialfrom> <speeddialto> </speeddialto></speeddialfrom></matchip></matchname></class></li> </ul>
- Display call waiting setting ATA24/voip/misc> callwait –s
<ul> <li>Edit call waiting setting ATA24/voip/sip&gt;callwait <port> <mode></mode></port></li> </ul>
- Display the codec setting ATA24/voip/sip> codec -s
<ul> <li>Edit prefect codec, codec rate and VAD for the port# ATA24/voip/sip&gt; codec <port> <prefercodec> <codecrate> <vad></vad></codecrate></prefercodec></port></li> </ul>
<ul> <li>Edit single codec for the port# ATA24/voip/sip&gt; codec –single <port> <active></active></port></li> </ul>
- Display VoIP setting ATA24/voip/sip>default_account -s
<ul> <li>Edit default SIP account</li> <li>ATA24/voip/sip&gt;default_account <port> <sip account=""></sip></port></li> </ul>
- Display dialplan setting ATA24/voip/sip>dialplan -s
- Display dialplan setting with detail description ATA24/voip/sip>dialplan -h
- Edit dialplan setting (adding new entry) ATA24/voip/sip>dialplan -a <matchstring> <minlength> <maxlength> <prefixstrip> <prefixadd> <sipipaddr> <interdigittimeout> <memo></memo></interdigittimeout></sipipaddr></prefixadd></prefixstrip></maxlength></minlength></matchstring>
- Edit dialplan setting (modifying an entry) ATA24/voip/sip>dialplan -e <entryidx> <matchstring> <minlength></minlength></matchstring></entryidx>
<maxlength><prefixstrip> <prefixadd> <sipipaddr> <interdigittimeout> <memo></memo></interdigittimeout></sipipaddr></prefixadd></prefixstrip></maxlength>
<interdigittimeout> <memo> - Delete dialplan setting ATA24/voip/sip&gt;dialplan -d <entryidx></entryidx></memo></interdigittimeout>
<pre><interdigittimeout> <memo> - Delete dialplan setting     ATA24/voip/sip&gt;dialplan -d <entryidx>     ATA24/voip/sip&gt;dialplan -D - Display DTMF Relay setting</entryidx></memo></interdigittimeout></pre>
<interdigittimeout> <memo> - Delete dialplan setting ATA24/voip/sip&gt;dialplan -d <entryidx> ATA24/voip/sip&gt;dialplan -D - Display DTMF Relay setting ATA24/voip/sip&gt;dtmf_relay -s - Edit DTMF relay mode for the port#</entryidx></memo></interdigittimeout>
InterDigitTimeOut> <memo> - Delete dialplan setting ATA24/voip/sip&gt;dialplan -d <entryidx> ATA24/voip/sip&gt;dialplan -D - Display DTMF Relay setting ATA24/voip/sip&gt;dtmf_relay -s - Edit DTMF relay mode for the port# ATA24/voip/sip&gt;dtmf_relay <port> <mode> - Edit DTMF relay mode and SIP INFO mode for the port#</mode></port></entryidx></memo>

# ATA24/voip/sip> fax -s

A 1 A 2 4 / V 0 1 p / s 1 p - 1 a x - s
- Edit fax mode for the port# ATA24/voip/sip> fax <port> <mode></mode></port>
- Display hotline setting ATA24/voip/sip> hotline -s
<ul> <li>Enable/Disable the hotline function</li> <li>ATA24/voip/sip&gt; hotline <port> <active></active></port></li> </ul>
<ul> <li>Edit the hotline number</li> <li>ATA24/voip/sip&gt; hotline <port> <active> &lt;<digits></digits></active></port></li> </ul>
- Display local listening port number for SIP ATA24/voip/sip> localport –s
<ul> <li>Edit SIP local port number ATA24/voip/sip&gt; localport <port></port></li> </ul>
- Display port activation setting ATA24/voip/sip> port_active -s
- Choose proxy for the port ATA24/voip/sip> port_active <port> <active></active></port>
- Display proxy server setting ATA24/voip/sip> server -s
<ul> <li>Enable/Disable the proxy server</li> <li>ATA24/voip/sip&gt; server <proxy#> <active></active></proxy#></li> </ul>
<ul> <li>Enable/Disable the proxy server and outbound proxy</li> <li>ATA24/voip/sip&gt; server <proxy#> <active> <outbound></outbound></active></proxy#></li> </ul>
- Edit the proxy server parameters ATA24/voip/sip> server <proxy#> <active> <outbound> <proxyname> <proxyip> <proxyport> <registrarip> <registrarport> <expires> <domain></domain></expires></registrarport></registrarip></proxyport></proxyip></proxyname></outbound></active></proxy#>
- Display SIP message (for SIP) ATA24/voip/sip> siplog <mode> ATA24/voip/sip&gt; siplog <mode><line></line></mode></mode>
- Display SIP user agent setting ATA24/voip/sip>sipua –s <index></index>
- Display ring port setting ATA24/voip/sip>sipua –r
<ul> <li>Edit SIP user agent setting</li> <li>ATA24/voip/sip&gt;sipua -e <index> <active> <username> <password></password></username></active></index></li> <li><displayname> <authid><callforwardmode> <callforwardurl></callforwardurl></callforwardmode></authid></displayname></li> <li><callforwardring><proxy> <callnoregister> <ringtype> <ipbind></ipbind></ringtype></callnoregister></proxy></callforwardring></li> </ul>
- Edit ring port setting ATA24/voip/sip> sipua –r <index> <ringport> <mode></mode></ringport></index>
- Delete SIP user agent setting ATA24/voip/sip> sipua -e ATA24/voip/sip>dialplan -D
- Display speed dial setting ATA24/voip/sip> speeddial –s

- Add speed dial number and destination for the entry ATA24/voip/sip> speeddial –a <number> <destination> <memo></memo></destination></number>	
<ul> <li>Edit speed dial number, destination and memo for the entry ATA24/voip/sip&gt; speeddial –e <index> <number> <destination> <memo></memo></destination></number></index></li> </ul>	
- Delete the entry of speed dial ATA24/voip/sip> speeddial –d <index></index>	
- Delete all entries of speed dial ATA24/voip/sip> speeddial –D	
- Display ports that unlocked ATA24/voip/sip> unlock –s ATA24/voip/sip> unlock –s <port></port>	
- Execute port unlock ATA24/voip/sip> unlock	<port></port>
<index></index>	1 to 30
	1 to 32 for SIP user agent
<name></name>	Name of the incoming calls
<ip domain=""></ip>	IP address or domain name
<class></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></matchname>	0 : Disable ; 1 : Enable
<matchip></matchip>	0 : Disable ; 1 : Enable
<speeddialfrom></speeddialfrom>	1 to 150
<speeddialto></speeddialto>	1 to 150
<port> &lt;</port>	Port number of the device.
	From 1 to 24
<mode></mode>	0 : Disable ; 1 : Enable
for ring port setting/RTP threshold	
setting	
<prefercodec></prefercodec>	Select one Codec to be applied on this port. IVD
	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></codecrate>	Select one rate value to be applied on this port.
< CoulerRate>	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></vad>	"Enable" to activate VAD(Voice Activity
	Detection, also known as Silence Suppression)
	function.
	"Disable" to stop using VAD. (Default is
	Disable)
<active></active>	"Enable" to activate this port.
	"Disable" to close this port. (Default is Disable)
	Disuble to crose this port. (Default is Disuble)

	1 to 22	
< <u>SIP Account&gt;</u>	1 to 32	
<pre><entryidx></entryidx></pre>		
<matchstring></matchstring>	Matched string, ex: 9011x.T, maximum 63 characters.	
<minlength></minlength>	Min. length of digits, range: 0~63, default: 0 (only	
	use for x.T (unfixed length))	
<maxlength></maxlength>	Max. length of digits, range: 0~63, default:32	
<prefixstrip></prefixstrip>	Number of prefix digits to strip, range: 0~63	
<prefixadd></prefixadd>	Prefix string to be add,	
	-1: none	
	maximum 63 char.	
<sipipaddr></sipipaddr>	SIP IP address or domain name, ex: iptel.org 0 for	
	no specific address	
<interdigittimeout></interdigittimeout>	Override the inter-digits timeout, range: 1~60(sec)	
	default: 4 (sec)	
<memo></memo>	User-specified name for comment, maximum 63	
	characters. Users can add some descriptions for	
	each number.	
	(Default is none)	
<mode></mode>	0: Disable	
	1: RFC2833 (Default is 1)	
	2: SIP INFO	
<sipinfomode></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)	
<pre><gain value=""></gain></pre>		
< 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></active>	0: Disable, 1: Enable	
	Or	
	0: off, 1: on	
<digits></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.	

< RegistrarIP > Assign an IP address or domain name of SIP register server. (Default is 0) < RegistrarPort > Assign a port number of SIP register server. (165535 (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> 0: Output last 50 lines for SIP Message 1: Output last N lines <line> Print last N lines for mode 1 for SIP Message <username> SIP username <password> SIP password <displayname> SIP authentication ID <callforwardmode> 0: Disable 1: Call forwarding all calls 2: Call forwarding no answer <callforwardurt> SIP url format, ex: 101@iptel.org  <callforwardring> 1-10 (rings) <proxy 2<="" p=""> 0: Don't use proxy server 1: use Proxy 1 2: use Proxy 2 3: use Proxy 2 3: use Proxy 3</proxy></callforwardring></callforwardurt></callforwardmode></displayname></password></username></line></mode></domain>		165535 (Default is 5060)	
register server. (Default is 0)         < RegistrarPort >       Assign a port number of SIP register server.         165335 (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>       0: Output last 50 lines         for SIP Message       1: Output last N lines         <username>       SIP username         <print 1<="" for="" last="" lines="" mode="" n="" td="">       for SIP Message         <username>       SIP password         <username>       SIP display name         <authid>       SIP authentication ID         <callforwardmode>       0: Disable         1: Call forwarding no answer       Call forwarding no answer         <callforwardring>       1-10 (rings)         <proxy>       0: Don't use proxy server         1: use Proxy 1       2: use Proxy 3         <username< td="">       0: Call with Registration         1: Call without Registration       1: Call without Registration         1: Call without Registration       1: Call without Registration         2: use Proxy 1       2: Rings by round robin          0: Rings all ports in the group         1: Rings the first available port<th>&lt; RegistrarIP &gt;</th><th></th></username<></proxy></callforwardring></callforwardmode></authid></username></username></print></username></mode></domain>	< RegistrarIP >		
165535 (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>       0: Output last 50 lines         for SIP Message       1: Output last N lines         <line>       Print last N lines for mode 1         for SIP Message       SIP username         <username>       SIP password         <displayname>       SIP display name         <authld>       SIP ausentication ID         <callforwardmode>       0: Disable         1: Call forwarding all calls       2: Call forwarding no answer         <callforwardurl>       SIP urf format, ex: 101@iptel.org         <callforwardring>       110 (rings)         <proxy>       0: Don't use proxy server         1: use Proxy 1       2: use Proxy 2         3: use Proxy 3       2: Call with Registration         <callnoregister>       0: Rings all ports in the group         1: Rings the first available port       2: Rings by round robin          1: VPN/LAN1         2: VPN/LAN3       4: VPN/LAN4          4: VPN/LAN3          124 port</callnoregister></proxy></callforwardring></callforwardurl></callforwardmode></authld></displayname></username></line></mode></domain>	5	-	
165535 (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>       0: Output last 50 lines         for SIP Message       1: Output last S0 lines          Print last N lines for mode 1         for SIP Message       SIP username          Vername&gt;          SIP display name          SIP display name          SIP display name          Authld&gt;         SIP display name           CallForwardMode&gt;         0: Disable       1: Call forwarding all calls         2: Call forwarding no answer          <callforwardurl>       SIP url format, ex: 101@iptel.org          CallForwardRing&gt;         1~10 (rings)          <proxy 1<="" td="">       2: use Proxy 1         2: use Proxy 2       3: use Proxy 3         <callnoregister>       0: Call with Registration         1: Call without Registration       1: Call without Registration         1: Call without Registration       1: Rings the firsta vailable port</callnoregister></proxy></callforwardurl></mode></domain>	< RegistrarPort >	Assign a port number of SIP register server.	
<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> (0: Output last 50 lines for SIP Message I: Output last N lines <line> Print last N lines for mode 1 for SIP Message  <username> SIP username <password> SIP assword  <th>5</th><th></th></password></username></line></mode></domain></expires>	5		
(minimum 60 seconds)          Assign an IP address or domain name of SIP         Domain/Realm. (Default is 0)          O: Output last 50 lines         for SIP Message       1: Output last N lines          Visername          Print last N lines for mode 1         for SIP Message       Print last N lines for mode 1          SIP username          Password>          SIP password          SIP display name <authid>       SIP authentication ID          CallForwardMode&gt;         0: Disable       1: Call forwarding no answer          CallForwardIrl&gt;         SIP url format, ex: 101@iptel.org          CallForwardRing&gt;         1~10 (rings)         <proxy< td="">       0: Don't use proxy server         1: use Proxy 1       2: use Proxy 3          Call with Registration         1: Call without Registration       1: Call without Registration          Call without Registration         1: Call without Registration       1: Call without Registration          Call without Registration          0: Rings all ports in the group      <tr< th=""><th>&lt; Expires &gt;</th><th colspan="2"></th></tr<></proxy<></authid>	< Expires >		
<domain>       Assign an IP address or domain name of SIP         Domain/Realm. (Default is 0)       O: Output last 50 lines         for SIP Message       1: Output last N lines         <line>       Print last N lines for mode 1         for SIP Message       Print last N lines for mode 1         <line>       Print last N lines for mode 1         for SIP Message       SIP username         <line>       Print last N lines for mode 1         for SIP Message       SIP username         <line>       SIP password         <displayname>       SIP password         <callforwardmode>       O: Disable         1: Call forwarding all calls       2: Call forwarding no answer         <callforwardring>       I~10 (rings)         <proxy>       O: Don't use proxy server         1: use Proxy 1       2: use Proxy 2         3: use Proxy 2       3: use Proxy 3         <callnoregister>       O: Rings all ports in the group         1: Rings the first available port       2: Rings by round robin         <ipsilow< td="">       0: WAN         VPN/LAN1       2: VPN/LAN3         VPN/LAN3       4: VPN/LAN4         <index>       1~24 port         <index>       1~24 port         <index></index></index></index></ipsilow<></callnoregister></proxy></callforwardring></callforwardmode></displayname></line></line></line></line></domain>	-	default value is 300.	
>Mode>       0: Output last 50 lines         for SIP Message       1: Output last 50 lines <line>       Print last N lines for mode 1         for SIP Message       -         <username>       SIP username         <password>       SIP password         <displayname>       SIP display name         <authid>       SIP authentication ID         <callforwardmode>       0: Disable         1: 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 1         2: use Proxy 2       3: use Proxy 3         <callnoregister>       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         <index>       1~150         for speed dial setting       -24 port</index></ipbind></callnoregister></proxy></callforwardring></callforwardurl></callforwardmode></authid></displayname></password></username></line>			
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for SIP Message       1: Output last N lines <line>       Print last N lines for mode 1         for SIP Message      </line>		Domain/Realm. (Default is 0)	
<ul> <li><line> Print last N lines for mode 1</line></li> <li>for SIP Message</li> <li></li> <li></li></ul>	<mode></mode>	0: Output last 50 lines	
for SIP Message <username>         SIP username         <password>         SIP assword         SIP assword         SIP assword         SIP authentication ID         <callforwardmode>         0: Disable         1: Call forwarding all calls         2: 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 with Registration         1: Call without Registration         4            2: Rings by round robin         4: RingType&gt;         0: WAN         1: VPN/LAN1         2: VPN/LAN2         3: VPN/LAN3         4: VPN/LAN4         <ringport>         1~24 port            4: VPN/LAN4         <ringport>         1~25 port         4: VPN/LAN4</ringport></ringport></callnoregister></proxy></callforwardring></callforwardurl></callforwardmode></password></username>	for SIP Message	1: Output last N lines	
<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       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 2       3: use Proxy 3         <callnoregister>       0: Call with Registration         1: Call without Registration       1: Call without Registration         1: Call without Registration       1: Call without Registration          1: Rings the first available port         2: Rings by round robin       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>       1~150         for speed dial setting       4.sign a dialing phone number.Ex: 101</index></ringport></ipbind></callnoregister></proxy></callforwardring></callforwardurl></callforwardmode></authid></displayname></password></username>	<line></line>	Print last N lines for mode 1	
<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       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       3: use Proxy 3         <callnoregister>       0: Call with Registration         1: Call without Registration       1: Call without Registration         1: Rings the first available port       2: Rings by round robin         <ip>VPN/LAN1       2: VPN/LAN1         2: VPN/LAN2       3: VPN/LAN3         4: VPN/LAN4       3: VPN/LAN4         <ringport>       1~24 port         <index>       1~150         for speed dial setting       Assign a dialing phone number.Ex: 101</index></ringport></ip></callnoregister></proxy></callforwardring></callforwardurl></callforwardmode></authid></displayname></password>	for SIP Message		
CDisplayName>       SIP display name <authid>       SIP authentication ID         <callforwardmode>       0: Disable         1: Call forwarding all calls       2: Call forwarding busy         3: Call forwarding no answer       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 2       3: use Proxy 3         <callnoregister>       0: Call with Registration         1: Call without Registration       1: Call without Registration         1: Rings all ports in the group       1: Rings the first available port         2: Rings by round robin       1: VPN/LAN1         2: VPN/LAN3       4: VPN/LAN3         4: VPN/LAN4       1~24 port         <index>       1~150         for speed dial setting       Assign a dialing phone number.Ex: 101</index></callnoregister></proxy></callforwardring></callforwardurl></callforwardmode></authid>	<username></username>	SIP username	
<authid>       SIP authentication ID         <callforwardmode>       0: Disable         1: Call forwarding all calls       2: Call forwarding busy         3: Call forwarding no answer       3: Call forwarding no answer         <callforwardurl>       SIP url format, ex: 101@iptel.org         <callforwardring>       1~10 (rings)         <callforwardring>       1~10 (rings)         <callforwardring>       0: Don't use proxy server         1: use Proxy 1       2: use Proxy 2         3: use Proxy 2       3: use Proxy 3         <callnoregister>       0: Call with Registration         1: Call without Registration       1: Call without Registration         1: Rings all ports in the group         1: Rings by round robin       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>       1~150         for speed dial setting       4xsign a dialing phone number.Ex: 101</index></ringport></ipbind></callnoregister></callforwardring></callforwardring></callforwardring></callforwardurl></callforwardmode></authid>	<password></password>	SIP password	
<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         1: Call without Registration         1: Call without Registration         1: 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         <index>         1~24 port         <index>         for speed dial setting         <number>       Assign a dialing phone number.Ex: 101</number></index></index></ipbind></callnoregister></proxy></callforwardring></callforwardurl></callforwardmode>	<displayname></displayname>	SIP display name	
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         1: Call without Registration         1: Call without Registration         1: 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         <index>         1~24 port         <index>         1~150         for speed dial setting            <number></number></index></index></ipbind></callnoregister></proxy></callforwardring></callforwardurl>	<authid></authid>	SIP authentication ID	
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         1: 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/LAN3         4: VPN/LAN4         <ringport>         1~24 port         <index>         for speed dial setting         <number></number></index></ringport></ipbind></callnoregister></proxy></callforwardring></callforwardurl>	<callforwardmode></callforwardmode>	0: Disable	
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       3: call with Registration          1: Call without Registration         1: RingType&gt;       0: Rings all ports in the group          2: Rings by round robin          2: VPN/LAN1         2: VPN/LAN3       4: VPN/LAN4         <ringport>       1~24 port         <index>       1~150         for speed dial setting       Assign a dialing phone number.Ex: 101</index></ringport></proxy></callforwardring></callforwardurl>		1: Call forwarding all calls	
<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       3: use Proxy 3         <callnoregister>       0: Call with Registration         1: Call without Registration       1: Call without Registration         1: Rings all ports in the group       1: Rings the first available port         2: Rings by round robin       2: VPN/LAN1         2: VPN/LAN1       2: VPN/LAN3         4: VPN/LAN4       3: VPN/LAN4         <index>       1~24 port         <index>       1~150         for speed dial setting       Assign a dialing phone number.Ex: 101</index></index></callnoregister></proxy></callforwardring></callforwardurl>		2: Call forwarding busy	
<callforwardring>       1~10 (rings)         <proxy>       0: Don't use proxy server         1: use Proxy 1       2: use Proxy 2         3: use Proxy 3       3: use Proxy 3         <callnoregister>       0: Call with Registration         1: Call without Registration       1: Call without Registration         <ringtype>       0: Rings all ports in the group         2: Rings by round robin       2: Rings by round robin         <ipbind>       0: WAN         1: VPN/LAN1       2: VPN/LAN2         3: VPN/LAN3       4: VPN/LAN4         <index>       1~24 port         <index>       1~150         for speed dial setting       Assign a dialing phone number.Ex: 101</index></index></ipbind></ringtype></callnoregister></proxy></callforwardring>			
<proxy> <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         1: Rings all ports in the group         1: Rings the first available port         2: Rings by round robin         <ip>VPN/LAN1         2: VPN/LAN2         3: VPN/LAN3         4: VPN/LAN4         <index>         for speed dial setting         <number></number></index></ip></callnoregister></proxy></proxy>	<callforwardurl></callforwardurl>	SIP url format, ex: 101@iptel.org	
1: use Proxy 1         2: use Proxy 2         3: use Proxy 3 <callnoregister>         0: Call with Registration         1: Call without Registration         0: Rings all ports in the group         1: Rings the first available port         2: Rings by round robin         0: WAN         1: VPN/LAN1         2: VPN/LAN2         3: VPN/LAN3         4: VPN/LAN4               Assign a dialing phone number.Ex: 101</callnoregister>	<callforwardring></callforwardring>		
2: use Proxy 2         3: use Proxy 3 <callnoregister>       0: Call with Registration         1: Call without Registration         1: Call without Registration         1: Call without Registration         <ringtype>         0: Rings all ports in the group         1: Rings the first available port         2: Rings by round robin         <ip>(PN/LAN1         2: VPN/LAN2         3: VPN/LAN3         4: VPN/LAN3         4: VPN/LAN4         <ringport>         1~24 port         <index>         for speed dial setting            <number>         Assign a dialing phone number.Ex: 101</number></index></ringport></ip></ringtype></callnoregister>	<proxy></proxy>		
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         <number>       Assign a dialing phone number.Ex: 101</number></index></ringport></ipbind></ringtype></callnoregister>			
<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         <number>       Assign a dialing phone number.Ex: 101</number></index></ringport></ipbind></ringtype></callnoregister>			
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         <number>         Assign a dialing phone number.Ex: 101</number></index></ringport></ipbind></ringtype>			
<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         <number>       Assign a dialing phone number.Ex: 101</number></index></ringport></ipbind></ringtype>	<callnoregister></callnoregister>		
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         <number>         Assign a dialing phone number.Ex: 101</number></index></ringport></ipbind>			
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         <number>         Assign a dialing phone number.Ex: 101</number></index></ringport></ipbind>	<ringtype></ringtype>		
<ipbind> 0: WAN 1: VPN/LAN1 2: VPN/LAN2 3: VPN/LAN3 4: VPN/LAN3 4: VPN/LAN4  Assign a dialing phone number.Ex: 101</ipbind>			
1: VPN/LAN1         2: VPN/LAN2         3: VPN/LAN3         4: VPN/LAN4 <ringport>         1~24 port         <index>         for speed dial setting         <number>         Assign a dialing phone number.Ex: 101</number></index></ringport>			
2: VPN/LAN2         3: VPN/LAN3         4: VPN/LAN4 <ringport>         1~24 port         <index>         for speed dial setting         <number>         Assign a dialing phone number.Ex: 101</number></index></ringport>	<ipbind></ipbind>		
3: VPN/LAN3         4: VPN/LAN4 <ringport>       1~24 port         <index>       1~150         for speed dial setting       Assign a dialing phone number.Ex: 101</index></ringport>			
4: VPN/LAN4 <ringport>       1~24 port         <index>       1~150         for speed dial setting          <number>       Assign a dialing phone number.Ex: 101</number></index></ringport>			
<ringport>       1~24 port         <index>       1~150         for speed dial setting          <number>       Assign a dialing phone number.Ex: 101</number></index></ringport>			
<index>     1~150       for speed dial setting     Assign a dialing phone number.Ex: 101</index>			
for speed dial setting		· · · · · · · · · · · · · · · · · · ·	
<number> Assign a dialing phone number.Ex: 101</number>		1~150	
	`¥¥		
<b><destination></destination></b> Assign an address of dialing destination. Ex:	<destination></destination>	<b>u</b>	
<u>101@iptel.org</u>		101@iptel.org	

# A.7.7 Statistics Commands

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

- Display call statistics setting ATA24/voip/statistics> callstat

- Display the setting by port ATA24/voip/statistics> callstat <Port>
- Edit the range for callstat port ATA24/voip/statistics> callstat <Port> <Range>
- Display RTP statistics setting ATA24/voip/statistics> rtpstat
- Display the setting by port ATA24/voip/statistics> rtpstat <Port>

- Edit the range for rtpstat port ATA24/voip/statistics> rtpstat <Port> <Range>

- Display RTP threshold setting ATA24/voip/statistics> rtpthreshold -s
- Edit the value for rtpthreshold

ATA24/voip/statistics> rtpthreshold <mode> <delayLow> <delayHigh> <jitterLow> <jitterHigh> <lostLow> <lostHigi> <timeout>

- Display VoIP RTP alert setting

ATA24/voip/statistics> showalert

#### - Display the setting by port

ATA24/voip/statistics> showalert <Port>

<port></port>	Port number of the device.	
	From 1 to 24	
<range></range>	0: 15 minutes	
for VoIP call statistics	1: 24 hour	
<delaylow></delaylow>	Round Trip Delay Low Threshold (ms)	
<delayhigh></delayhigh>	Round Trip Delay High Threshold (ms)	
<jitterlow></jitterlow>	Jitter Low Threshold (ms)	
<jitterhigh></jitterhigh>	Jitter High Threshold (ms)	
<lostlow></lostlow>	Packet Loss Ratio Low Threshold (0100%)	
<losthigh></losthigh>	Packet Loss Ratio High Threshold (0100%)	
<timeout></timeout>	RTCP timeout (in seconds)	
<lowfreq></lowfreq>	Assign a low frequency number in Hertz unit.	
	(unit is HZ) (Default is 350)	
<highfreq></highfreq>	Assign a high frequency number in Hertz unit.	
	(unit is HZ) (Default is 440)	
<ton1></ton1>	The duration of the first ringing.	
	(10msec per unit) (Default is 0)	
<toff1></toff1>	The silence duration after the first ringing.	
	(10msec per unit) (Default is 0)	
<ton2></ton2>	The duration of the next continuous ringing.	
	(10msec per unit) (Default is 0)	
<toff2></toff2>	The silence duration after the next continuous	
	ringing.	
	(10msec per unit) (Default is 0)	
<type></type>	0: North America	
for call ID setting	1: JAPAN	
	2: ETSI (Default is 2)	
	3: DTMF	

# A.7.8 VoIP Status Commands

- Help in the Statistics function ATA24/voip/status> ?	
- Display VoIP faults ATA24/voip/status>faultstatus	
<ul> <li>Display VoIP FXS port hook state (onhook or offhook) ATA24/voip/status&gt;hookstate ATA24/voip/status&gt;hookstate<port></port></li> </ul>	
<ul> <li>Display VoIP connection Status</li> <li>ATA24/voip/status&gt;portstatus</li> <li>ATA24/voip/status&gt;portstatus <port></port></li> </ul>	
<ul> <li>Display VoIP SIP User Agent Registration Status ATA24/voip/status&gt;sipuastatus ATA24/voip/status&gt;sipuastatus <port></port></li> </ul>	
<ul> <li>Display VoIP Status</li> <li>ATA24/voip/status&gt;voipstatus</li> <li>ATA24/voip/status&gt;voipstatus <mode></mode></li> </ul>	
<port></port>	Port number of the device. From 1 to 24
<mode></mode>	0: disable

## A.7.9 Tone User Defined Commands

- Help in the Statistics function ATA24/voip/tone/user defined>?
- Display user defined tone setting ATA24/voip/tone/user\_defined> busy -s
- Edit frequency and cadence for busy tone ATA24/voip/tone/user\_defined> busy <Lowfreq> <Highfreq> <Ton1> <Toff1> <Ton2> <Toff2>

1: enable

- Display caller ID setting ATA24/voip/tone/user\_defined> callerid -s
- Edit caller id type

ATA24/voip/tone/user\_defined> callerid <Type>

- Display the setting ATA24/voip/tone/user\_defined> congestion -s
- Edit frequency and cadence for congestion tone ATA24/voip/tone/user\_defined> congestion <Lowfreq> <Highfreq> <Ton1> <Toff1> <Ton2> <Toff2>
- Display user defined dial tone setting ATA24/voip/tone/user\_defined> dial -s
- Edit frequency and cadence for dial tone ATA24/voip/tone/user\_defined> dial <Lowfreq> <Highfreq> <Ton1> <Toff1> <Ton2> <Toff2>
- Display user defined ringing tone setting ATA24/voip/tone/user\_defined> ringing -s

#### - Edit frequency and cadence for ringing tone

ATA24/voip/tone/user\_defined> ringing <Lowfreq> <Highfreq> <Ton1> <Toff1> <Ton2> <Toff2>

- Display the country of the tone setting ATA24/voip/tone> region -s
- Choose the region for CPT setting ATA24/voip/tone> region <Region Number>

- Display CPT tone timer setting ATA24/voip/tone/ timer -s

- Edit CPT tone timer

ATA24/voip/tone/timer <Tone> <Timer>

<lowfreq></lowfreq>	(unit is HZ) (Default is 440)	
<highfreq></highfreq>	(unit is HZ) (Default is 440) (units is HZ) (Default is 480)	
<ton1></ton1>	(10msec per unit) (Default is 0)	
<toff1></toff1>	(10msec per unit) (Default is 0)	
<ton1> <ton2></ton2></ton1>		
	(10msec per unit) (Default is 200)	
<toff2></toff2>	(10msec per unit) (Default is 400)	
for user defined ring tone		
<region number=""></region>	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></tone>	1: Dial Tone	
for CPT tone timer setting	2: Busy Tone	
C	3: Howler Tone	
	4: Ringing Tone	
	5: Special Dial Tone	
	6: Call waiting Tone	
	7: Congestion Tone	
	8: Reorder Tone	
<timer></timer>	Range: 0~300 <sec></sec>	

# A.7.10 Config Commands

- Help

ATA24/voip>protocol ?

- Execute/activate VoIP setting

ATA24/voip>config

# A.7.11 List Commands

- Help

ATA24/voip>listcmds ?

- Display all VoIP CLI commands ATA24/voip>listcmds

# A.7.12 Protocol Commands

- Help

ATA24/voip>protocol ?

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

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