



User Guide FGW4148-16S/32S/48S

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About This User Guide

Thanks for choosing FGW4148-16S/32S/48S router with VoIP. This product will allow you to make ATA call using your broadband connection.

This manual provides basic information on how to install and connect FGW4148-16S/32S/48S router with VoIP to the Internet. It also includes features and functions of router with VoIP components, and how to use it correctly.

Before you can connect FGW4148-16S/32S/48S to the Internet and use it, you must have a high-speed broadband connection installed. A high-speed connection includes environments such as DSL, cable modem, and a leased line.

FGW4148-16S/32S/48S router with VoIP is a stand-alone device, which requires no PC to make Internet calls. This product guarantees clear and reliable voice quality on Internet, which is fully compatible with SIP industry standard and able to interoperate with many other SIP devices and software on the market.







- Chapter 1Product description
- Chapter 2 IVR Voice Prompt
- Chapter 3 Basic Settings
- Chapter 4Web Interface
- Chapter 5Troubleshooting Guide



Contacting FlyingVoice

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Purpose

The documents are intended to instruct and assist personnel in the operation, installation and maintenance of the FlyingVoice equipment and ancillary devices. It is recommended that all personnel engaged in such activities be properly trained. FlyingVoice disclaims all liability whatsoever, implied or express, for any risk of damage, loss or reduction in system performance arising directly or indirectly out of the failure of the customer, or anyone acting on the customer's behalf, to abide by the instructions, system parameters, or recommendations made in this document.

Cross references

References to external publications are shown in italics. Other cross references, emphasized in blue text in electronic versions, are active links to the references.

This document is divided into numbered chapters that are divided into sections. Sections are not numbered, but are individually named at the top of each page, and are listed in the table of contents.

Feedback

We appreciate feedback from the users of our documents. This includes feedback on the structure, content, accuracy, or completeness of our documents. Send feedback to support@flyingvoice.com.

Declaration of Conformity

Part 15 FCC Rules

This device complies with Part 15 of the FCC Rules. Operation is subject to the following two conditions:

- This device may not cause harmful interference, and
- This device must accept any interference received, including interference that may cause undesired operation.

Class B Digital Device or Peripheral

This equipment has been tested and found to comply with the limits for a Class B digital device, pursuant to Part 15 of the FCC Rules. These limits are designed to provide reasonable protection against harmful interference in a residential installation. This equipment can generate, use and radiate radio frequency energy. If not installed and used in accordance with the instruction manual, may cause harmful interference to radio communications. However, there is no guarantee that interference does not occur in a particular installation.



Notes

Changes or modifications not expressly approved by the party responsible for compliance could void the user's authority to operate the equipment.

If this equipment does cause harmful interference to radio or television reception, which can be determined by turning the equipment off and on, the user is encouraged to try to correct the interferences by one or more of the following measures:

- Reorient or relocate the receiving antenna.
- Increase the separation between the equipment and receiver.
- Connect the equipment into an outlet on a circuit different from that to which the receiver is connected.
- Consult the dealer or an experienced radio/TV technician for help.

Warnings and Notes

The following describes how warnings and notes are used in this document and in all documents of the FlyingVoice document set.

Warnings

Warnings precede instructions that contain potentially hazardous situations. Warnings are used to alert the reader to possible hazards that could cause loss of life or physical injury. A warning has the following format:



Warning

Warning text and consequence for not following the instructions in the warning.

Notes

A note means that there is a possibility of an undesirable situation or provides additional information to help the reader understand a topic or concept. A note has the following format:



Notes

Notes text and consequence for not following the instructions in the Notes.

Chapter 1 Product description

This chapter covers:

- FGW4148-16S/32S/48S
- LED Indicators and Interfaces
- Hardware Installation

FGW4148-16S/32S/48S

Table 1 Features at-a-glance

Port/Model	FWG4148-16S	FWG4148-32S	FWG4148-48S
------------	-------------	-------------	-------------

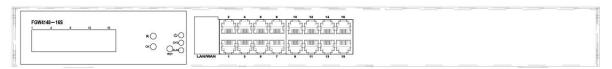


Ethernet	2*RJ45 10/100/1000M	2*RJ45 10/100/1000M	2*RJ45 10/100/1000M
interface	(WAN/LAN)	(WAN/LAN)	(WAN/LAN)
FXS	16	32	48
SIP Account	16	32	48
Wire-speed NAT	Support	Support	Support
DHCP	Client/Server	Client/Server	Client/Server
Voice Code	G.711 (A-law, U-law), G.729A/AB,G.72	3,G.722
Management	Voice menu, Web Mar	nagement, Provision:TFTP/HTTF	P/HTTPS, TR069, SNMP
Fax		T.30, T.38 Fax	

LED Indicators and Interfaces

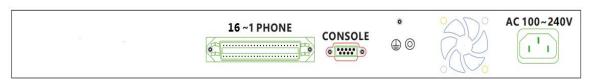
FGW4148-16S

Table 2 FGW4148-16S Front panel



LED/Interface	Status	Description
LED		
LED Screen	The screen will appe	ear the information about the device
450	On(Green)	The router is powered on and running normally.
	Off	The router is powered off.
eve	Blinking(Green)	The router is updating
3130	on	The router is running normally.
ALM	When the light is on	, it indicates system applications occured
Ethernet interface	Blinking(Green)	Have data transmission
indicator	on	The system is not powered on or the network port is not connected to the network device
Interface		
*	Page up,can view	info of FXS status.
ок	Use OK to return to	the standby page
O _A	Press it to restore fa	actory settings above 5S
LAN	Connector for local i	network devices
WAN	Connector for access	sing the internet
PHONE 1-16	Connect to the phor	ne

Table 3 FGW4148-16S Rear panel

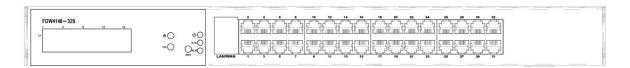


Interface	Description
16-1 PHONE	Transfer interface for 1-16 FXS
CONSOLE	Serial port,used to connect to your PC
Ground	Connect the ground wire
Fan	For equipment cooling

AC 100~220V	Connector for a power adapter.	

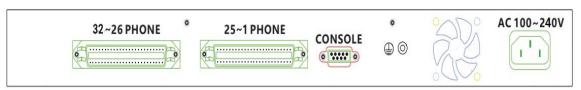
FGW4148-32S

Table 4 FGW4148-32S Front panel



LED/Interface	Status	Description	
LED			
LED Screen	The screen will appea	The screen will appear the information about the device	
d) O	On(Green)	The router is powered on and running normally.	
	Off	The router is powered off.	
eve	Blinking(Green)	The router is updating	
3130	on	The router is running normally.	
ALM	When the light is on, it indicates system applications occured		
Ethernet interface	Blinking(Green)	Have data transmission	
indicator	on	The system is not powered on or the network port is not connected to the network device	
Interface			
*	Page up, can view info of FXS status.		
ок	Use OK to return to the standby page		
O _A RST	Press it to restore factory settings above 5S		
LAN	Connector for local network devices		
WAN	Connector for accessing the internet		
PHONE 1-16	Connect to the phone		

Table 5 FGW4148-32S Rear panel

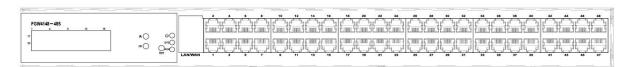


Interface	Description	
32-26 PHONE	Transfer interface for 26-32 FXS	
25-1 PHONE	Transfer interface for 1-25 FXS	
CONSOLE	Serial port,used to connect to your PC	

Ground	Connect the ground wire	
Fan	For equipment cooling	
AC 100~220V	Connector for a power adapter.	

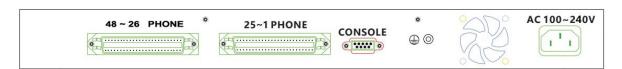
FGW4148-48S

Table 6 FGW4148-48S Front panel



LED/Interface	Status	Description
LED		
LED Screen	The screen will appea	r the information about the device
d) O	On(Green)	The router is powered on and running normally.
	Off	The router is powered off.
sys	Blinking(Green)	The router is updating
	on	The router is running normally.
ALM	When the light is on,	it indicates system applications occured
Ethernet interface	Blinking(Green)	Have data transmission
indicator	on	The system is not powered on or the network port is not connected to the network device
Interface		
* ○	Page up, can view info of FXS status.	
ок	Use OK to return to the standby page	
O _A RST	Press it to restore factory settings above 5S	
LAN	Connector for local network devices	
WAN	Connector for accessing the internet	
PHONE 1-16	IE 1-16 Connect to the phone	

Table 7 FGW4148-48S Rear panel



Interface	Description
48-26 PHONE	Transfer interface for 26-48 FXS
25-1 PHONE	Transfer interface for 1-25 FXS

CONSOLE	Serial port,used to connect to your PC	
Ground	Connect the ground wire	
Fan	For equipment cooling	
AC 100~220V	Connector for a power adapter.	

Hardware Installation

Before configuring your router, please see the procedure below for instructions on connecting the device in your network.

Procedure 1 Configuring the Router

- 1. Connect analog phone to ATA Port with an RJ11 cable.
- 2. Connect the WAN port to the Interne your network's modem/switch/router/ADSL
- 3. equipment using an Ethernet cable.
- 4. Connect one end of the power cord to the power port of the device. Connect the other end to the wall outlet.
- 5. Check the Power, WAN, and LAN LED to confirm network connectivity.

Warning



Please do not attempt to use unsupported power adapters and do not remove power during configuring or updating the device. Using other power adapters may damage

FGW4148-16S/32S/48S and will void the manufacturer warranty.

Warning



Changes or modifications not expressly approved by the party responsible for compliance can void the user's authority to operate the equipment.

This equipment has been tested and found to comply with the limits for a Class B digital device, pursuant to Part 15 of the FCC Rules. These limits are designed to provide reasonable protection against harmful interference in a residential installation. This equipment generates, uses and can radiate radio frequency cause harmful interference to radio communications. However, there is no energy and, if not installed and used in accordance with the instructions, may guarantee that interference will not occur in a particular installation.

If this equipment does cause harmful interference to radio or television reception, which can be determined by turning the equipment off and on, the user is encouraged to try to correct the interference by one or more of the following measures:

- Reorient or relocate the receiving antenna.
- Increase the separation between the equipment and receiver.
- Connect the equipment into an outlet on a circuit different from that to which the receiver is connected.

Chapter 2 IVR Voice Prompt

This chapter contains:

- Voice adapter Configuration Method (IVR)
- IVR description

Voice adapter Configuration Method (IVR)

The device can be configured in two ways, as follows:

- (1) Use IVR (Interactive Voice Response)
- (2) the use of web pages

This chapter mainly introduces how to configure the voice adapter through IVR.

Start IVR

Users follow these steps to achieve IVR:

- (1) Go off-hook and press the "****" key to start the IVR. Then the user will hear the voice prompt "1 WAN port configuration...".
- (2) According to different options, press any digit between 0 and 9, the device will broadcast the corresponding content, the numbers 0 to 9 represent the details as shown in the chart below.
 - (3) After each setting is successful, the device will play "Please input option, 1 WAN port configuration...".



Note

Before using IVR, please confirm analog phone is connected with ATA correctly.

IVR Description

The following chart lists the IVR requirements and a detailed description:

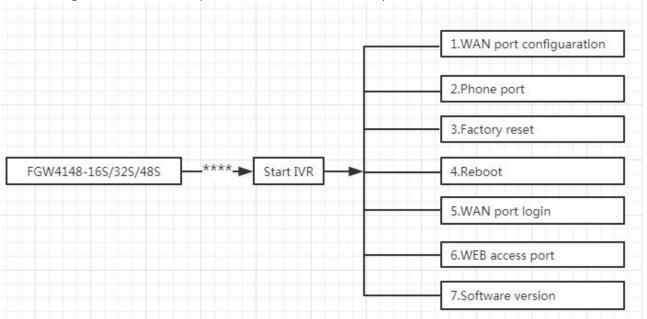


Table 8 IVR Menu Setting Options

Operation	Menu
Operation code 1 (1) WAN Port Configuration	 Pick up phone and press "****" to start IVR Select "1", then the device will continue to broadcast to remind users to choose 1.WAN port connection type; 2.WAN port IP address; 3. WAN subnet mask; 4. adapter; 5. DNS Choose "1", and The router reports the current WAN port connection type2) Prompt "Please enter password", user needs to input password and press "#" key, if user wants to configuration WAN port connection type. The password in IVR is same as web management interface login, the user may use phone keypad to enter password directly For example: WEB login password is "admin", so the password in IVR is "admin". The user may "23646" to access and then configure the WAN connection port. The unit reports "Operation Successful" if the password is correct.
	 Prompt "Please enter password", user needs to input password and press "#" key if user wants to configuration WAN port connection type. Choose the new WAN port connection type (1) DHCP or (2) Static The unit reports "Operation Successful" if the changes are successful. The router returns to the prompt "please enter your option" To quit, enter "*"

	1. Pick up phone and press "****" to start IVR	
	2. Choose "2", and The router reports current WAN Port IP Address	
	3. Input the new WAN port IP address and press "#" key:	
	4. Use "*" to replace ".", for exampleuser can input 192*168*20*168 to set	
(2)	the new IP address 192.168.20.168	
WAN Port IP	5. Press # key to indicate that you have finished	
Address	6. Report "operation successful" if user operation is ok.	
	7. To quit, enter "**".	
	1. Pick up phone and press "****" to start IVR	
	2. Choose "3", and router reports current WAN port subnet mask	
	3. Input a new WAN port subnet mask and press # key:	
(3)	4. Use "*" to replace ".", user can input 255*255*255*0 to set the new	
WAN Port	WAN port subnet mask 255.255.255.0	
Subnet Mask	5. Press "#" key to indicate that you have finished	
	6. Report "operation successful" if user operation is ok.	
	7. To quit, enter "**".	
	1. Pick up phone and press "****" to start IVR	
	2. Choose "4", and the router reports current adapter	
(4) adapter	3. Input the new adapter and press "#" key:	
	4. Use "*" to replace ".", user can input 192*168*20*1 to set the new	
	adapter 192.168.20.1.	
	5. Press "#" key to indicate that you have finished.	
	6. Report "operation successful" if user operation is ok.	
	7. To quit, press "**".	

	1. Pick up phone and press "***" to start IVR		
	2. Choose "5", and the router reports current DNS		
(5)	3. Input the new DNS and press # key:		
DNS	4. Use "*" to replace ".", user can input 192*168*20*1 to set the new		
	adapter 192.168.20.1.		
	5. Press "#" key to indicate that you have finished.		
2	1. Pick up phone and press "***" to start IVR		
phone port	2. Select "2", then the device will continue to broadcast prompts the user to select		
configuration	current phone number; 2. registration server address; 3. registration port; 4. call		
	forwarding configuration, 5. DNS configuration;		
	3. Continue pressing "1" and the unit will continue to broadcast the phone number of		
	the current phone port. The device will then broadcast "1. Phone number" again		
	1. Pick up phone and press "****" to start IVR		
	2. Choose "6", and the router reports "Factory Reset"		
3	3. Prompt "Please enter password", the method of inputting password is the same		
Factory Reset	actory Reset as operation 1.		
4. If you want to quit, press "*".			
	5. Prompt "operation successful" if password is right and then the router will be		
	Pick up phone and press "****" to start IVR		
	2. Choose "7", and the router reports "Reboot"		
4	3. Prompt "Please enter password", the method of inputting password is same as		
Reboot	operation 1.		
	4. the router reboots if password is right and operation		

	1. Pick up phone and press "****" to start IVR
5	2. Choose "8", and the router reports "WAN Port Login"
WAN Port Login	3. Prompt "Please enter password", the method of inputting password is same as
	operation 1.
	4. If user wants to quit, press "*".
	1. Pick up phone and press "****" to start IVR
6	2. Choose "9", and the router reports " WEB Access Port"
WEB Access	3. Prompt "Please enter password", the method of inputting password is same as
Port	operation 1.
	4. Report "operation successful" if user operation is ok.
7	1. Pick up phone and press "****" to start IVR
Firmware Version	2. Choose "0" and the router reports the current Firmware version



Note

- 1. While using Voice menu, press * (star) to return to main menu.
- 2.If any changes made in the IP assignment mode, the router must be rebooted in order for the settings to take effect.
- 3. While entering an IP address or subnet mask, use "*" (star) to enter "." (Dot) and use "#" (hash) key to finish entering IP address or subnet mask:
- 4.For example, to enter the IP address 192.168.20.159 by keypad, press these keys: 192*168*20*159, use the #(hash) key to indicate that you have finished entering the IP address.
- 5.Use the # (hash) key to indicate that you have finish entering the IP address or subnet mask
- 6.While assigning an IP address in Static IP mode, setting the IP address, subnet mask and default adapter is required to complete the configuration. If in DHCP mode, please make sure that a DHCP server is available in your existing broadband connection to which WAN port of FGW4148-16S/32S/48S is connected.
- 7.The default LAN port IP address of FGW4148-16S/32S/48S is 192.168.11.1 and this address should not be assigned to the WAN port IP address of FGW4148-16S/32S/48S in the same network segment of LAN port.
- 8. The password can be entered using phone keypad, the mapping table between number and letters as follows:

```
To input: D, E, F, d, e, f -- press '3'

To input: G, H, I, g, h, i -- press '4'

To input: J, K, L, j, k, I -- press '5'

To input: M, N, O, m, n, o -- press '6'

To input: P, Q, R, S, p, q, r, s -- press '7'

To input: T, U, V, t, u, v -- press '8'

To input: W, X, Y, Z, w, x, y, z -- press '9'
```

To input all other characters in the administrator password----press '0'.

Chapter 3 Basic Settings

This chapter covers:

- WEB Page
- SIP Account Register
- Basic Function

WEB Page

About Password

Our device supports two levels of management: administrators and users.

- (1) Administrator mode can browse and set all configuration parameters.
- (2) User mode can set all configuration parameters except SIP1/2 that some parameters can not be changed, such as server address and port.
 - Default user with administrator mode: Username: admin, Password: admin
 - Default user with user mode: Username: admin, Password: user

URL Format

FGW4148-16S/32S/48S has a built-in web server in response to HTTP get / post requests. Users can use a web browser, such as Microsoft's IE, to log in to the FGW4148-16S/32S/48S page and configure the FGW4148-16S/32S/48S

LAN port Login

1.Ensure your PC is connected to the router's LAN port correctly.



Note

You may either set up your PC to get an IP dynamically from the router or set up the IP address of the PC to be the same subnet as the default IP address of router is 192.168.1.1. For detailed information, see Chapter 5: Troubleshooting Guide.

- 2.Open a web browser on your PC and input "http://192.168.1.1".
- 3. The following window appears and prompts for username, password.



- 4.For administrator mode operation, please type admin/admin on Username/Password and click Login to begin configuration.
- 5. For user mode operation, please type user/user on Username/Password and click Login to begin configuration.

A

Note

If you are unable to access the web configuration, please see Chapter 5: Troubleshooting Guide for more information.

6.The web management interface automatically logs out the user after 5 minutes of inactivity.

WAN port Login

- 1.Ensure your PC is connected to the router's WAN port correctly.
- 2.Obtain the IP addresses of WAN port using Voice prompt or by logging into the device web management interface via a LAN port and navigating to Network > WAN.
- 3.Open a web browser on your PC and input http://<IP address of WAN port>. The following login page will be opened to enter username and password.



- 4. For administrator mode operation, type admin/admin on Username/Password and click Login to begin configuration.
- 5. For user mode operation, type user/user on Username/Password and click Login to begin configuration.

Note

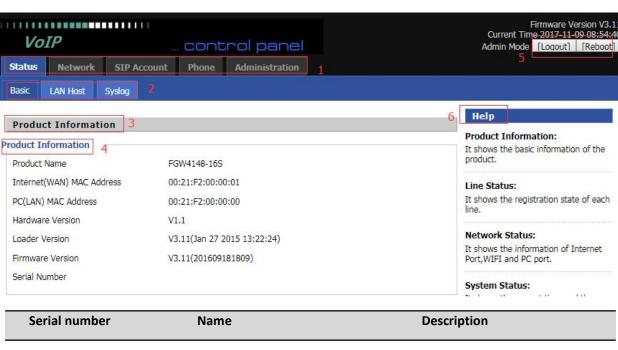


If you fail to access to the web configuration, see Chapter 6: Troubleshooting Guide for more information.

6. The web management interface automatically logs out the user after 5 minutes of inactivity.

WEB Interface Introduction

Table 9 WEB Interface Introduction



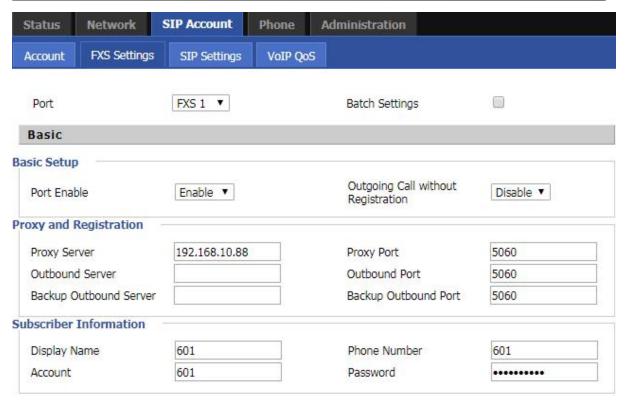
Serial number	Name	Description
Postition 1	navigation bar	Click navigation bar, many sub-navigation bar will appear in the place 2
Postition 2	sub-navigation bar	Click sub-navigation bar to enter to configuration page
Postition 3	configuration title	The configuration title
Postition 4	configuration bars	The configuration bars
Postition 5	main information	Display the firmware version, DSP version, Current Time, and user can change login level (mode) to return to login page by press blue Switch button.
Postition 6	Help	Display the main information for configuration; user can get help from it directly.
	Save	After changing the parameters, you need to click this button to save. After you click Save, there is a need to restart the device.
	Cancel	Click to cancel the change
	Reboot	Click to restart
	Refresh	Refresh current page

SIP Account Register

FGW4148-16S/32S/48S have 16/32/48 Lines to make SIP (Session Initiation Protocol) calls. Before registering, the device user should have SIP accounts configured by the system administrator or provider. See the section below for more information.

Register one by one

Table 10 Config SIP the Web Management Interface



Steps:

- Step 1. The account enable is set to "On" and the line can be used after opening.
- Step 2. The registration server fills in the IP address of the SIP server.
- Step 3. Display Name Fill in the content is the name of the number displayed on the LCD.
- Step 4. The registration account is filled with the account provided by the SIP server.
- Step 5. The name of the authentication is the SIP account provided by the SIP server.
- Step 6. The password is filled with the password provided by the SIP server registration account.
- Step 7. When you are finished, click the Save button at the bottom of the page to make the configuration take effect.
- Step 8. Check the registration of the corresponding line on the display / web status page.



Notes

Step 3-9 is to fill in the required content, other parameters fill in the required

Procedure

To view the SIP account status of device, open the **Status** web page and view the value of registration status.

Batch Registration

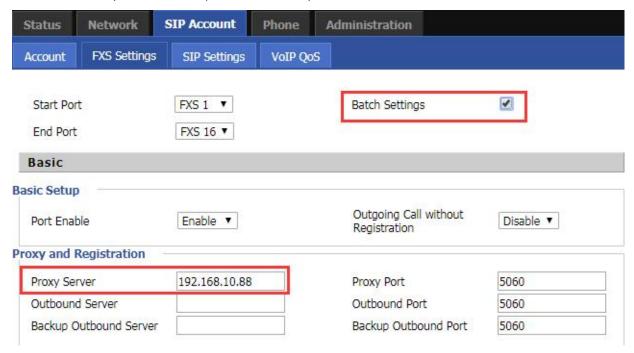
There are many FXS ports on the FGW4148-16S / 32S / 48S. One by one, configuration is very troublesome. Therefore, we support batch configuration of SIP accounts.

Taking FGW4148-16S as an example, batch configuration of SIP account steps:

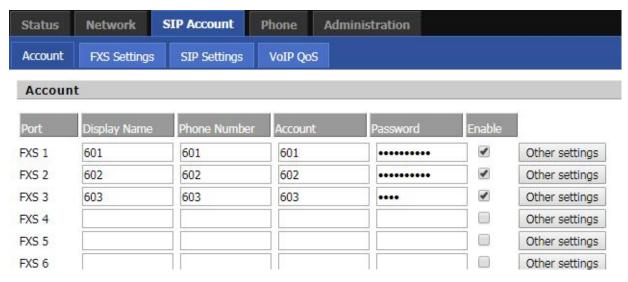
1. Log in to the web page, switch to the SIP Account - FXS Settings page, check the "Batch Settings", and select

the need to set the batch FXS port.

2. Fill in the "Proxy Server", other parameters on request.



3.Switch to SIP Account - Account page, fill in the batch configuration of FXS port account as required.



- 4. Click "Save" button
- 5. Status page can view the registration status information.

Basic Function

Calling phone or extension numbers

To make a phone or extension number call:

- Both ATA and the other VoIP device (i.e., another ATA or other SIP products) must have public IP addresses, or
- Both ATA and the other VoIP device (i.e., another ATA or other SIP products) are on the same LAN
 using private or public IP addresses, or
- Both ATA and the other VoIP device (i.e., another ATA or other SIP products) can be connected through
 a router using a public or private IP addresses.

To make a call, first pick up the analog phone or turn on the speakerphone on the analog phone, input the IP address directly, end with #.

Direct IP calls

Direct IP calling allows two phones, that is, an ATA with an analog phone and another VoIP Device, to talk to each other without a SIP proxy. VoIP calls can be made between two phones if:

- Both ATA and the other VoIP device (i.e., another ATA or other SIP products) have public IP addresses,
 or
- Both ATA and the other VoIP device (i.e., another ATA or other SIP products) are on the same LAN

using private or public IP addresses, or

Both ATA and the other VoIP device (i.e., another ATA or other SIP products) can be connected through
a router using public or private IP addresses.

To make a direct IP call, first pick up the analog phone or turn on the speakerphone on the analog phone, Input the IP address directly, with the end "#".

Call Hold

Method 1: While in conversation, pressing the "*77" to put the remote end on hold, then you will hear the dial tone and the remote party will hear hold tone at the same time.

Pressing the "*77" again to release the previously hold state and resume the bi-directional media.

Method 2: While in conversation, pressing the Flash key to put the remote end on hold, then you will hear the dial tone and the remote party will hear hold tone at the same time.

Pressing the Flash key again to release the previously hold state and resume the bi-directional media.

Call transfer

1.Blind Transfer

Assume that call party A and party B are in conversation. Party A wants to Blind Transfer B to C:

Method 1: Party A dials "*98" to get a dial tone, then dials party C's number, and then press immediately key # (or wait for 4 seconds) to dial out. A can hang up.

Method 2: Party A pressed the Flash key to get a dial tone, then dials party C's number, and then press immediately key # (or wait for 4 seconds) to dial out. A can hang up. C answers the call and talks to B.

When enabled R key and Flash as R key function, you can refer to the steps below to transfer:

Party A pressed the Flash key and number 4 key to get a dial tone, then dials party C's number, and then press immediately key # (or wait for 4 seconds) to dial out. A can hang up. C answers the call and talks to B.

2.Attended Transfer

Assume that call party A and B are in a conversation. A wants to Attend Transfer B to C:

Method 1: Party A dials "*77" to hold the party B, when hear the dial tone, A dials C's number, then party A and party C are in conversation.

Party A dials "*98" to transfer to C, then B and C now in conversation.

If the transfer is not completed successfully, then A and B are in conversation again.

Method 2: Party A presses the Flash key to hold the party B, when hear the dial tone, A dials C's

number, then party A and party C are in conversation. Party A hang up, then B and C now in conversation.

Conference

Assume that call party A and B are in a conversation. A wants to add C to the conference:

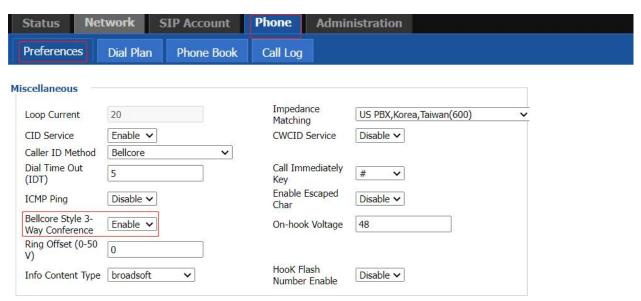
Method 1: Party A dials "*77" to hold the party B, when hear the dial tone, A dial C's number, then party A and party C are in conversation.

Party A dials "*88" to add C, then A and B, for conference.

Method 2: Party A pressed Flash key to hold the party B, when hear the dial tone, A dial C's number, then party A and party C are in conversation.

Party A pressed Flash key again, then A, B and C are in the same conversation.

Note: If this method is not working, please enable Bellcore Style 3-Way Conference option on the web interface.



Chapter 4 Web Interface

This chapter guides users to execute advanced (full) configuration through admin mode operation. This chapter covers:

- Login
- Status
- Network
- SIP Account
- Phone
- Administration

Login

Table 11 Login details

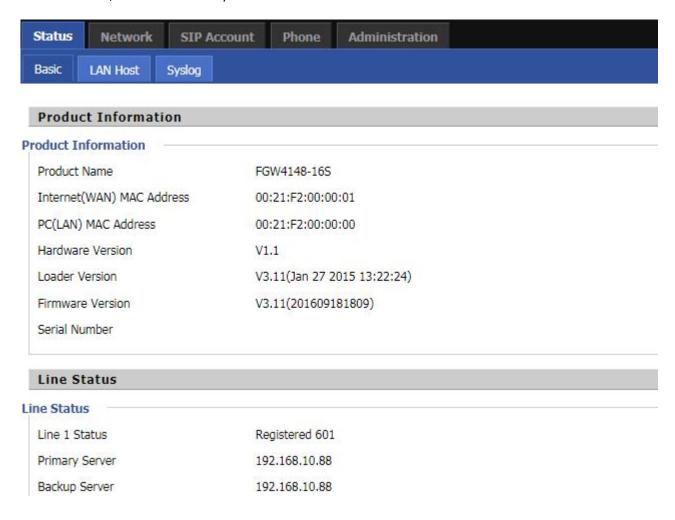


Procedure

- 1. Connect the LAN port of the router to your PC an Ethernet cable
- 2. Open a web browser on your PC and type http://192.168.1.1.
- 3. Enter Username admin and Password admin.
- 4. Click Login

Status

This webpage shows the status information about the Product, Network, SIP Account Status, FXS Port Status, Network Status, Wireless Info and System Status



Network

You can configure the WAN port, LAN port, DDNS, Multi WAN, DMZ, MAC Clone, Port Forward and other parameters in this section of the web management interface.

WAN

This page allows you to set WAN configuration with different modes. Use the Connection Type drop down list to choose one WAN mode and then the corresponding page will be displayed.

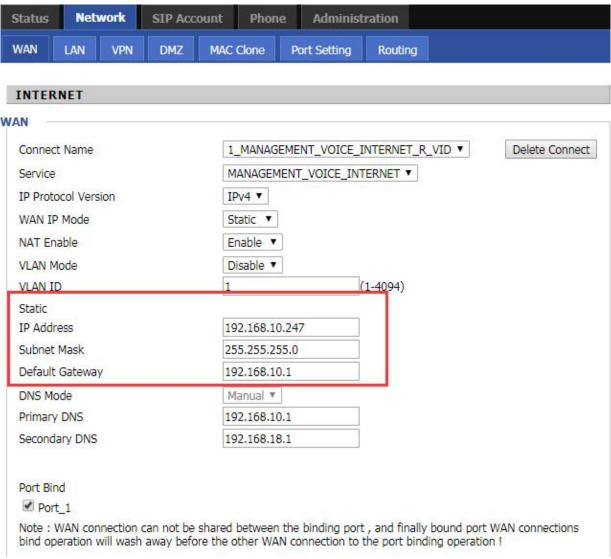
1.Static IP

This configuration may be utilized when a user receives a fixed public IP address or a public subnet,

namely multiple public IP addresses from the Internet providers. In most cases, a Cable service provider will offer a fixed public IP, while a DSL service provider will offer a public subnet. If you have a public

subnet, you can assign an IP address to the WAN interface.

Table 12 Static IP



Field Name	Description	
IP Address	The IP address of Internet port	
Subnet Mask	The subnet mask of Internet port	
Default adapter	The default adapter of Internet port	
	Select DNS mode, options are Auto and Manual:	
	1. When DNS mode is Auto, the device under LAN port will	
DNS Mode	automatically obtain the preferred DNS and alternate DNS.	
	2. When DNS mode is Manual, the user manually configures the	
	preferred DNS and alternate DNS information	

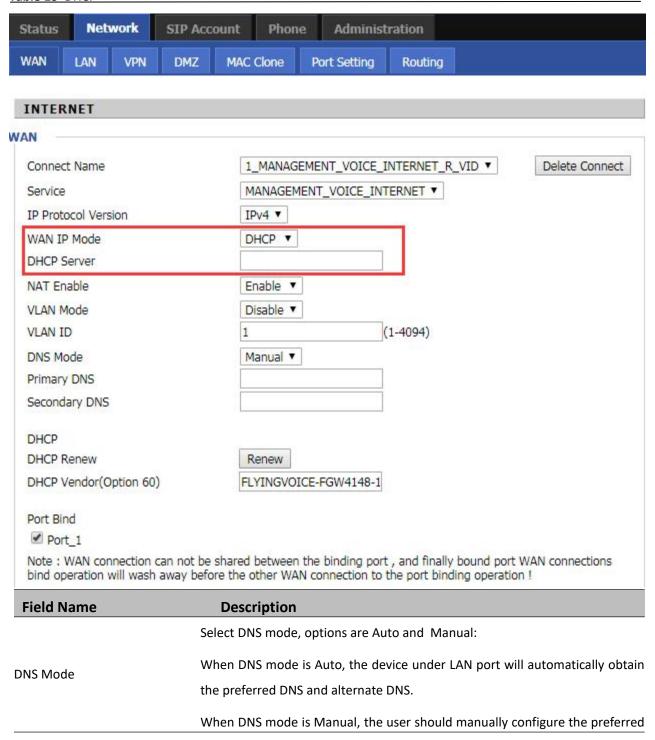
Primary DNS Address	The primary DNS of Internet port
Secondary DNS Address	The secondary DNS of Internet port

2.DHCP

The Router has a built-in DHCP server that assigns private IP address to each local client.

The DHCP feature allows to the router to obtain an IP address automatically from a DHCP server. In this case, it is not necessary to assign an IP address to the client manually.

Table 13 DHCP



Primary DNS Address	Primary DNS of Internet port.
Secondary DNS Address	Secondary DNS of Internet port.
DHCP Renew	Refresh the DHCP IP address
DHCP Vendor (Option60)	Specify the DHCP Vendor field. Display the vendor and product name.

3.PPPoE

PPPoE stands for Point-to-Point Protocol over Ethernet. It relies on two widely accepted standards: PPP and Ethernet. It connects users through an Ethernet to the Internet with a common broadband medium, such as a single DSL line, wireless device or cable modem. All the users over the Ethernet can share a common connection.

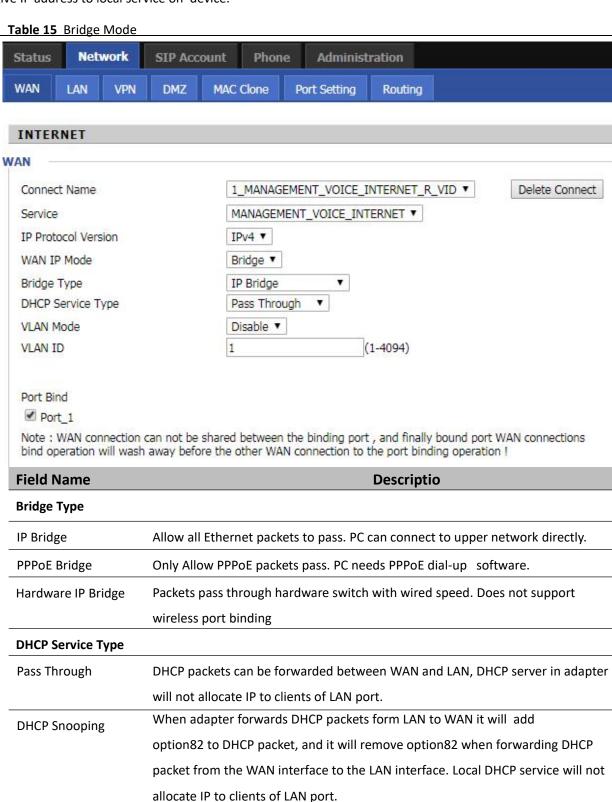
PPPoE is used for most of DSL modem users. All local users can share one PPPoE connection for accessing the Internet. Your service provider will provide you information about user name, password, and authentication mode.

Table 14 PPPoE Network SIP Account Status Phone Administration WAN LAN VPN DMZ MAC Clone Port Setting Routing INTERNET WAN Connect Name 1_MANAGEMENT_VOICE_INTERNET_R_VID ▼ Delete Connect Service MANAGEMENT_VOICE_INTERNET ▼ IP Protocol Version IPv4 ▼ WAN IP Mode PPPoE ▼ NAT Enable Enable ▼ VLAN Mode Disable ▼ VLAN ID (1-4094)• DNS Mode Auto Primary DNS Secondary DNS PPPoE PPPoE Account PPPoE Password Confirm Password Service Name Leave empty to autodetect Operation Mode Keep Alive Keep Alive Redial Period(0-3600s)

Field Name	Description
PPPoE Account	Enter a valid user name provided by the ISP
	Enter a valid password provided by the ISP. The password can contain special
PPPoE Password	characters and allowed special characters are $\$$, $+$, $*$, $\#$, $@$ and $!$ For example, the
	password can be entered as #net123@IT!\$+*.
Confirm Password	Enter your PPPoE password again
Service Name	Enter a service name for PPPoE authentication.
	If it is left emply, the service name is auto detected.
Operation Mode	Select the mode of operation, options are Keep Alive, On Demand and Manual:
	When the mode is Keep Alive, the user sets the 'keep alive redial period' values range
	from 0 to 3600s, the default setting is 5 minutes;
	When the mode is On Demand, the user sets the 'on demand idle time' value in the
	range of 0-60 minutes, the default setting is 5 minutes;
	Operation Mode On Demand
	On Demand Idle Time(0-60m) 5
Keep Alive Redial	Set the interval to send Keep Alive messaging
Period	
PPPoE Account	Assign a valid user name provided by the ISP

4.Bridge Mode

Bridge Mode under Multi WAN is different with traditional bridge setting. Bridge mode employs no IP addressing and the device operates as a bridge between the WAN port and the LAN port. Route Connection has to be built to give IP address to local service on device.



adapter will not forward DHCP packets between LAN and WAN, it also blocks	
DHCP packets from the WAN port. Clients connected to the LAN port can get IP	
from DHCP server run in adapter.	
The WAN interface is untagged. LAN is untagged.	
The WAN interface is tagged. LAN is untagged.	
Only valid in bridge mode. All ports, including WAN and LAN, belong to this VLAN	
Id and all ports are tagged with this VLAN id. Tagged packets can pass through	
WAN and LAN.	
Set the VLAN ID.	
Set the priority of VLAN, Options are 0~7.	



Note

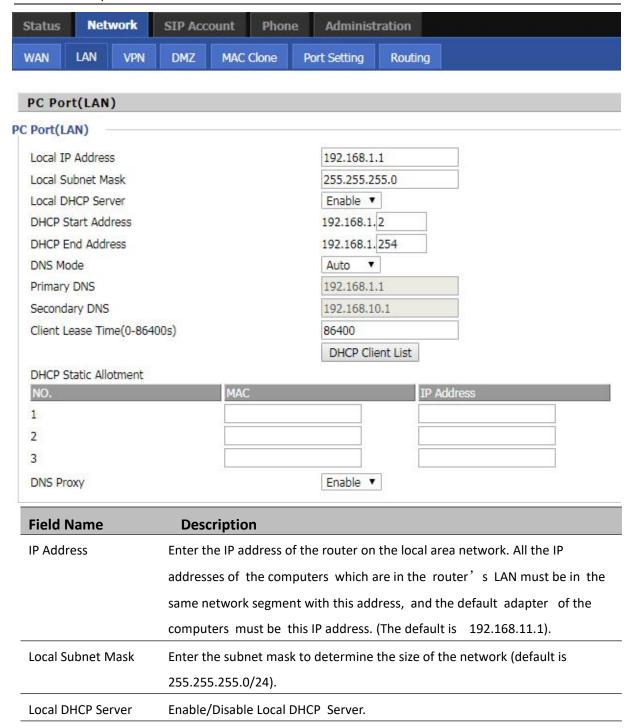
Multiple WAN connections may be created with the same VLAN ID

LAN

LAN Port

NAT translates the packets from public IP address to local IP address to forward packets to the proper destination.

Table 16 LAN port

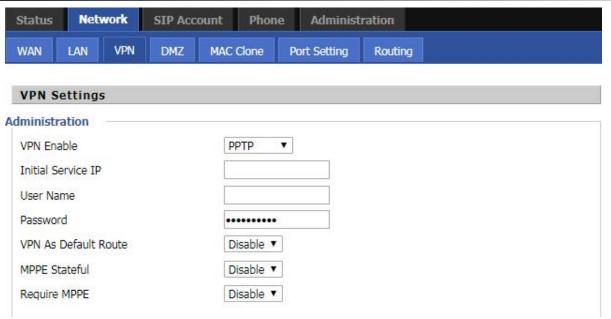


DHCP Start Address	Enter a valid IP address as a starting IP address of the DHCP server, and if the
	router's LAN IP address is 192.168.11.1, starting IP address can be
	192.168.11.2 or greater, but should be less than the ending IP address.
DHCP End Address	Enter a valid IP address as an end IP address of the DHCP server.
DNS Mode	Select DNS mode, options are Auto and Manual:
	When DNS mode is Auto, the device under LAN port will automatically obtains
	the preferred DNS and alternate DNS.
	When DNS mode is Manual, the user should manually configure the preferred
	DNS and alternate DNS.
Primary DNS	Enter the preferred DNS address.
Secondary DNS	Enter the secondary DNS address.
Client Lease Time	This option defines how long the address will be assigned to the computer
	within the network. In that period, the server does not assign the IP address to
	the other computer.
DNS Proxy	Enable or disable; If enabled, the device will forward the DNS request of LAN-
	side network to the WAN side network.

VPN

VPN is a technology that builds a private network on a public network. The connection between any two nodes of the VPN network does not have the end-to-end physical link required by the traditional private network, but rather the network platform provided by the public network service provider, and the user data is transmitted in the logical link. With VPN technology, you can establish private connections and transfer data between any two devices on the public network.

Table 17 PPTP



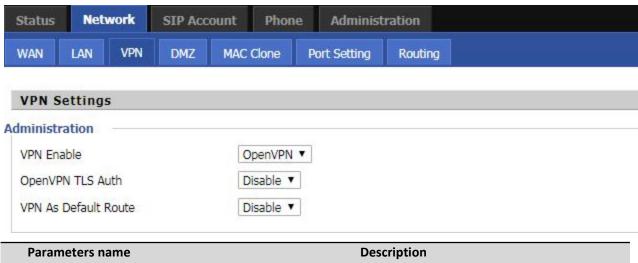
Parameters name	Description
VPN Enable	Whether to enable VPN. Select PPTP mode.
Initial Service IP	The IP address of the VPN server.
User Name	The user name required for authentication.
Password	The password required for authentication.
VPN As Default Route	Prohibited or open, the default is prohibited.
MPPE Stateful	Disable or enable MPPE Stateful.
Require MPPE	Disable or enable Require MPPE.

Table 18 L2TP



Parameters name	Description
VPN Fnable	Whether to enable VPN.
	Select PPTP mode.
Initial Service IP	The IP address of the VPN server.
User Name	The user name required for authentication.
Password	The password required for authentication.
L2TP Tunnel Name	L2TP Tunnel Name
L2TP Tunnel Password	L2TP Tunnel Password
VPN As Default Route	Prohibited or open, the default is prohibited.

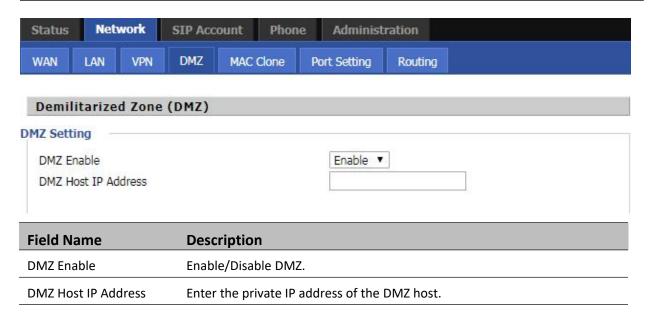
Table 19 OpenVPN



Parameters name	Description
VPN Enable	Whether to enable VPN.
	Select OpenVPN mode.
OpenVPN TLS Auth	Whether OpenVPN TLS authentication is enabled
VPN As Default Route	Prohibited or open, the default is prohibited.

DMZ

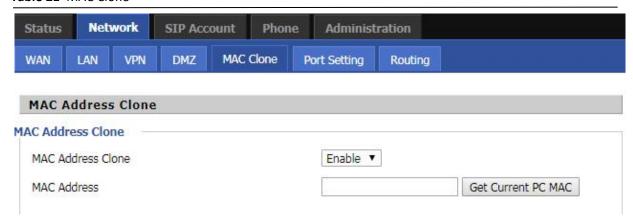
Table 20 DMZ



MAC Clone

Some ISPs will require you to register your MAC address. If you do not wish to re-register your MAC address, you can have the router clone the MAC address that is registered with your ISP. To use the Clone Address button, the computer viewing the Web-base utility screen will have the MAC address automatically entered in the Clone WAN MAC field.

Table 21 MAC Clone

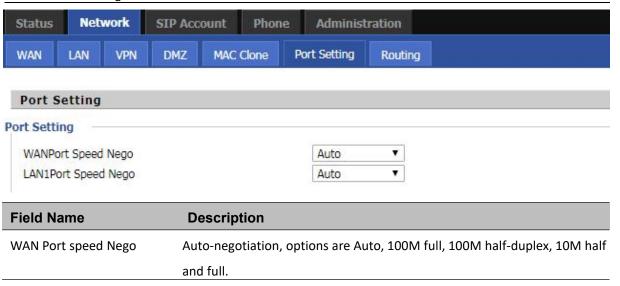


Config steps:

- 1. Enabling MAC address cloning
- 2. Press the button Get Current PC MAC gets PC's MAC address
- 3. Press the button Save to save your changes if users don't want to use MAC clone, press the button to cancel the changes
- 4. Press the button Reboot to make the changes effective.

Port Setting

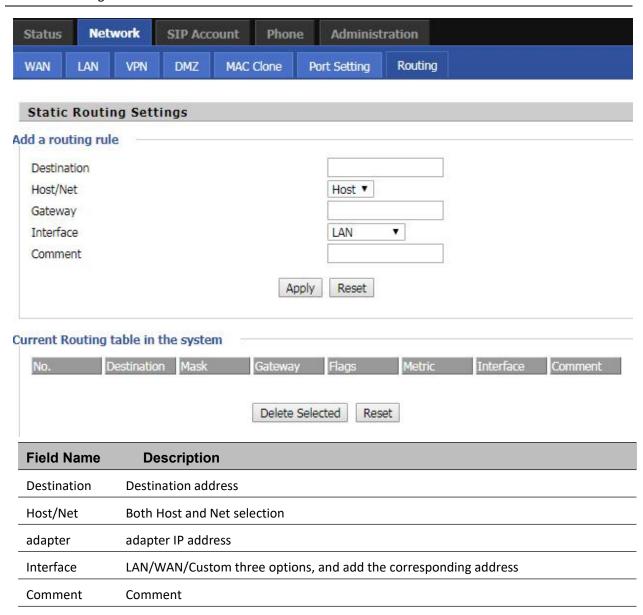
Table 22 Port setting



LAN Port Speed Nego	Auto-negotiation, options are Auto, 100M full, 100M half, 10M half and
	10M full.

Routing

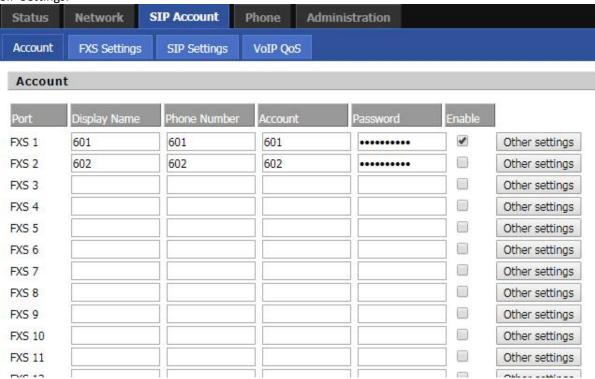
Table 23 Routing



SIP Account

Account

You can set each FXS's display name, phone number, account and password in this page, the corresponding FXS will be enable after checked enable .then save your settings. Click "Other settings" go to the "FXS Settings" web page. SIP Settings.



FXS Settings

Basic

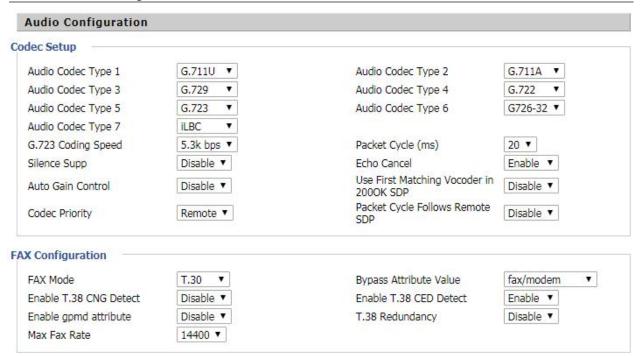
Set the basic information provided by your VOIP Service Provider, such as Phone Number, Account, password, SIP Proxy and others.

Table 24 Line Status Network SIP Account Phone Administration **FXS Settings** Account SIP Settings VoIP QoS Port FXS 1 ▼ Batch Settings Basic **Basic Setup** Outgoing Call without Port Enable Enable ▼ Disable ▼ Registration Proxy and Registration Proxy Server 192.168.10.88 Proxy Port 5060 Outbound Server Outbound Port 5060 Backup Outbound Server Backup Outbound Port 5060 Subscriber Information 601 Phone Number 601 Display Name 601 Password Account ********* **Field Name Description** Line Enable Enable/Disable the line. Enable/Disable Outgoing Call without Registration **Outgoing Call without** If enabled, SIP-1 will not send register request to SIP server; but in Status/SIP Registration Account Status webpage, Status is Registered; lines 1 can dial out, but the external line number cannot dialed line1. **Proxy Server** The IP address or the domain of SIP Server **Outbound Server** The IP address or the domain of Outbound Server Backup Outbound Server The IP address or the domain of Backup Outbound Server SIP Service port, default is 5060 Proxy port Outbound Port Outbound Proxy's Service port, default is 5060 **Backup Outbound Port** Backup Outbound Proxy's Service port, default is 5060 The number will be displayed on LCD Display Name **Phone Number** Enter telephone number provided by SIP Proxy Account Enter SIP account provided by SIP Proxy

Password Enter SIP password provided by SIP Proxy
rd Enter SIP password provided by SIP Pro

Audio Configuration

Table 25 Audio configuration



Field Name	Description
Audio Codec Type1	Choose the audio codec type from G.711U, G.711A, G.722, G.729, G.723
Audio Codec Type2	Choose the audio codec type from G.711U, G.711A, G.722, G.729, G.723
Audio Codec Type3	Choose the audio codec type from G.711U, G.711A, G.722, G.729, G.723
Audio Codec Type4	Choose the audio codec type from G.711U, G.711A, G.722, G.729, G.723
Audio Codec Type5	Choose the audio codec type from G.711U, G.711A, G.722, G.729, G.723
G.723 Coding Speed	Choose the speed of G.723 from 5.3kbps and 6.3kbps
Packet Cycle	The RTP packet cycle time, default is 20ms
Silence Supp	Enable/Disable silence support
Echo Cancel	Enable/Disable echo cancel. By default, it is enabled
Auto Gain Control	Enable/Disable auto gain
T.38 Enable	Enable/Disable T.38
T.38 Redundancy	Enable/Disable T.38 Redundancy
T.38 CNG Detect Enable	Enable/Disable T.38 CNG Detect
gpmd attribute Enable	Enable/Disable gpmd attribute

Supplementary Service Subscription

Table 26 Supplementary service

Supplementary Service Subscription			
Supplementary Services Call Waiting MWI Enable MWI Subscribe Enable DND	Enable ▼ Enable ▼ Disable ▼ Disable ▼	Hot Line Voice Mailbox Numbers VMWI Serv	Enable ▼
Speed Dial			
Speed Dial 2 Speed Dial 4 Speed Dial 6 Speed Dial 8		Speed Dial 3 Speed Dial 5 Speed Dial 7 Speed Dial 9	
Field Name	Description		
Call Waiting	Enable/Disable Call Waiting		
Hot Line	Fill in the hotline number, Pickup handset or press hands-free or headset button, the device will dial out the hotline number automatically		
MWI Enable	Enable/Disable MWI (message waiting indicate). If the user needs to user voice mail, please enable this feature		
MWI Subscribe Enable	Enable/Disable MWI Subscribe		
Voice Mailbox Numbers	Fill in the voice mailbox phone number, Asterisk platform, for example, its default voice mail is *97		
DND	Enable/Disable DND (do not disturb)		
Speed Dial	Enter the speed dial phone numbers. Dial *74 to active speed dial function Then press the speed dial numbers, for example, press 2, phone dials 075526099365 directly		

Advanced

Table 27 Advanced

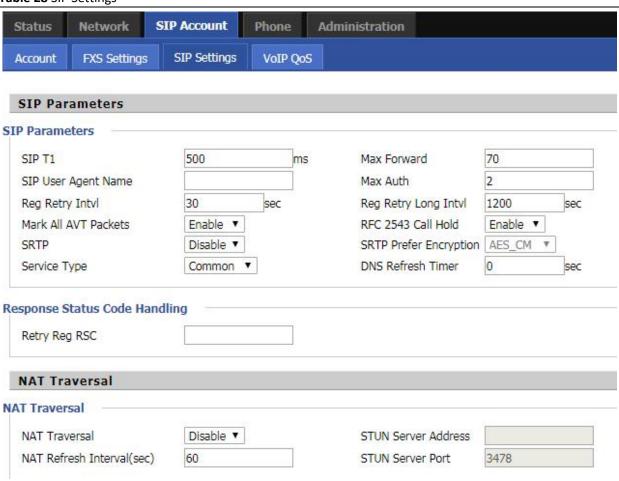
Advanced			
SIP Advanced Setup			
Domain Name Type	Enable ▼	Carry Port Information	Disable ▼
Signal Port	53378	DTMF Type	Inband ▼
RFC2833 Payload(>=96)	101	Register Refresh Interval(sec)	3600
Caller ID Header	FROM	▼ Remove Last Reg	Enable ▼
Session Refresh Time(sec)	0	Refresher	UAC ▼
SIP 100REL Enable	Disable ▼	SIP OPTIONS Enable	Disable ▼
Initial Reg With Authorization	Disable ▼	Reply 182 On Call Waiting	Disable ▼
Primary Server Detect Interval	0	Max Detect Fail Count	3
NAT Keep-alive Interval(10- 60s)	15	Anonymous Call	Disable ▼
Anonymous Call Block	Disable ▼	Proxy DNS Type	A Type ▼
Use OB Proxy In Dialog	Disable ▼	Complete Register	Disable ▼
Reg Subscribe Enable	Disable ▼	Reg Subscribe Interval(se	ec) 0
Dial Prefix		User Type	Phone ▼
Hold Method	ReINVITE	▼ Request-URI User Check	Enable ▼
Only Recv Request From Server	Disable ▼	Server Address	
SIP Received Detection	Disable ▼	VPN	Disable ▼
SIP Encrypt Type	Disable ▼	RTP Encrypt Type	Disable ▼
Country Code		Remove Country Code	Disable ▼
Tel URL	Disable ▼	Use Random SIP Port	Enable ▼
Min Random SIP Port	50000	Max Random SIP Port	60000
Prefer Primary SIP Server	Disable ▼		
RTP Advanced Setup	2		
RTP Port Min	0 (0 means a	auto select) RTP Port Max	50000
Parameter name	D	Description	
Domain Name Type	٧	Whether to enable domain name recognition	n in SIP URIs
Carry Port Information		Whether to carry the SIP URI port information	
Signal Port		The local port number of the SIP protocol	
DTMF Type		Select the second way of dialing, option RFC2833 and SIP Info.	nal items are In-band,
RFC2833 Payload(>=96)		The user can use the default settings	
Register Refresh Interval(sec)		The time interval between two normal registration messages. The user can use the default settings.	

Remove Last Reg Whether to remove the last registration message Session Refresh Time(sec) Refresher Select Refresh from UAC and UAS If this option is enabled, the IP phone will send SIP-OPTION to the server instead of sending Hello messages on a regular basis. The interval for sending is the parameter set for the "NAT Hold Interval" parameter. SIP OPTIONS Enable Whether to open the SIP OPTION function Initial Reg With Authorization Whether to open the SIP OPTION function Initial Reg With Authorization Whether to open the SIP OPTION function Mether anonymous Call waiting NAT Keep-alive Interval(10-60s) The time interval for sending empty packets Anonymous Call Anonymous Call Whether anonymous calls are enabled Anonymous Call Block Whether to enable anonymous call blocking Proxy DNS Type Set the DNS server type, the optional items are Type A, DNS SRV, and Auto Use OB Proxy In Dialog Whether the OB agent is used in the conversation Complete Register Whether to enable full registration Reg Subscribe Enable Reg Subscribe Interval(sec) Disable or enable Reg Subscription message is sent after the registration message; the subscription message is not sent when disabled Reg Subscribe Interval(sec) Dial before prefix User Type Whether the end user is IP or Phone Hold Method Call hold is REINVITE or INFO Request-URI User Check Whether to allow the user to check Only Recv Request From Server SIP server address SIP Received Detection Whether to allow SIP message encryption RTP Encrypt Type Whether to allow RTP message encryption Country Code Country Code	Caller ID Header	When enabled, an unregistered message will be sent before the registration is disabled, and no unregistered messages will be sent before registration; should be set according to the different server requirements
Refresher Select Refresh from UAC and UAS If this option is enabled, the IP phone will send SIP-OPTION to the server instead of sending Hello messages on a regular basis. The interval for sending is the parameter set for the "NAT Hold Interval" parameter. SIP OPTIONS Enable Whether to open the SIP OPTION function Initial Reg With Authorization Whether to carry the certification information when registering Whether or not to send 182 when the call is waiting NAT Keep-alive Interval(10-60s) The time interval for sending empty packets Anonymous Call Waiting Whether or not to send 182 when the call is waiting NAT Keep-alive Interval(10-60s) The time interval for sending empty packets Anonymous Call Whether anonymous calls are enabled Anonymous Call Block Whether to enable anonymous call blocking Proxy DNS Type Set the DNS server type, the optional items are Type A, DNS SRV, and Auto Use OB Proxy In Dialog Whether the OB agent is used in the conversation Complete Register Whether to enable full registration When enabled, the subscription message is sent after the registration message; the subscription message is not sent when disabled Reg Subscribe Interval(sec) Disable or enable Reg Subscribe Interval Dial Prefix Dial before prefix User Type Whether the end user is IP or Phone Hold Method Call hold is REINVITE or INFO Request-URI User Check Whether to allow the user to check Only Recv Request From Server other requests Server Address SIP server address SIP server address SIP server address SIP server address SIP server address SIP received Detection Whether to allow SIP message encryption RTP Encrypt Type Whether to allow SIP message encryption	Remove Last Reg	Whether to remove the last registration message
If this option is enabled, the IP phone will send SIP-OPTION to the server instead of sending Hello messages on a regular basis. The interval for sending is the parameter set for the "NAT Hold Interval" parameter. SIP OPTIONS Enable Whether to open the SIP OPTION function Initial Reg With Authorization Whether to carry the certification information when registering Reply 182 On Call Waiting Whether or not to send 182 when the call is waiting NAT Keep-alive Interval(10-60s) The time interval for sending empty packets Anonymous Call Whether anonymous calls are enabled Anonymous Call Block Whether to enable anonymous call blocking Proxy DNS Type Set the DNS server type, the optional items are Type A, DNS SRV, and Auto Use OB Proxy In Dialog Whether the OB agent is used in the conversation Complete Register Whether to enable full registration When enabled, the subscription message is sent after the registration message; the subscription message is not sent when disabled Reg Subscribe Interval(sec) Disable or enable Reg Subscribe Interval Dial Prefix Dial before prefix User Type Whether the end user is IP or Phone Hold Method Call hold is REINVITE or INFO Request-URI User Check Whether to allow the user to check Only Recv Request From Server of the server, do not accept other requests SIP server address SIP server address SIP Received Detection Whether to allow SIP receive detection VPN Whether to enable VPN SIP Encrypt Type Whether to allow RTP message encryption	Session Refresh Time(sec)	·
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RTP Encrypt Type Whether to allow RTP message encryption	VPN	Whether to enable VPN
	SIP Encrypt Type	Whether to allow SIP message encryption
Country Code Country code	RTP Encrypt Type	Whether to allow RTP message encryption
	Country Code	Country code

Remove Country Code	Whether to allow the removal of national codes
Tel URL	Whether to open the Tel URL
Use Random SIP Port	Whether to use the minimum random port
Min Random SIP Port	SIP minimum random port
Max Random SIP Port	SIP maximum random port
Prefer Primary SIP Server	Whether to enable the preferred primary server
Hold SDP Attribute Inactive	Whether to enable the call to keep the inactive attribute
RTP Port Min	RTP minimum port
RTP Port Max	RTP's maximum port

SIP Settings

Table 28 SIP Settings

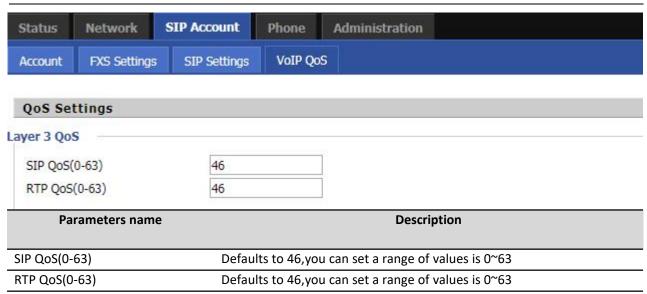


Parameters name	Description
SIP Parameters	
SIP T1	The default value is 500
SIP User Agent Name	Enter the SIP User Agent header field
Max Forward	Modify the maximum hop value, the default is 70

Max Auth	Change the number of authentication failures, the default value is 2
Reg Retry Intvl	Registration failed again registration interval, default is 30
Reg Retry Long Intvl	Registration failed Register again for the long interval Default 1200
Mark All AVT Packets	The default enable is on
RFC 2543 Call Hold	The default enable is on
SRTP	The default is disabled
SRTP Prefer Encryption	Support for AES_CM and ARIA_CM
Service Type	Default general
DNS Refresh Timer	Modify the DNS refresh time, the default value of 0
Transport	The transmission type defaults to UDP
Response Status Code Handling	
Retry Reg RSC	You can fill in Retry Reg RSC here
NAT Traversal	
NAT Traversal	Whether to enable NAT mode, or select STUN to penetrate
STUN Server Address	STUN server IP address
NAT Refresh Interval(sec)	Refresh interval
STUN Server Port	STUN port, the default is 3478

VoIP QoS

Table 29 VoIP QoS



Configuration can be based on the scene environment to modify the parameters

Phone

Preferences

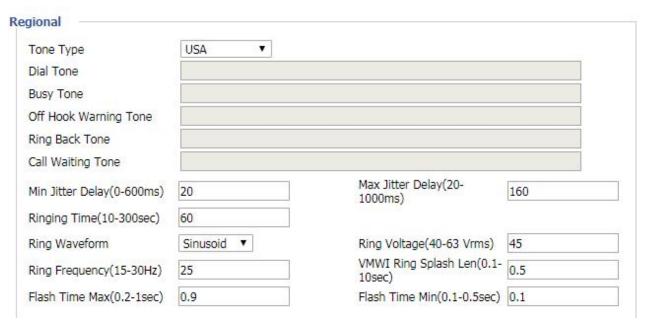
Preferences

Table 30 Preferences



Regional

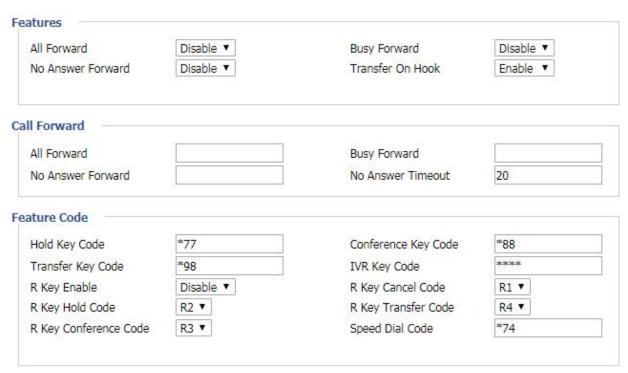
Table 31 Regional



Field Name	Description
Tone Type	Choose tone type form China, US, Hong Kong and so on.
Dial Tone	Dial Tone
Busy Tone	Busy Tone
Off Hook Warning	Off Hook warning tone
Ring Back Tone	Ring back tone
Call Waiting Tone	Call waiting tone
Min Jitter Delay	The Min value of home adapter's jitter delay, home adapter is an adaptive jitter
	mechanism.
Max Jitter Delay	The Max value of home adapter's jitter delay, home adapter is an adaptive jitter
	mechanism.
Ringing Time	How long CnPilot Home R190/R200x will ring when there is an incoming call.
Ring Waveform	Select regional ring waveform, options are Sinusoid and Trapezoid, the default
	Sinusoid.
Ring Voltage	Set ringing voltage, the default value is 70
Ring Frequency	Set ring frequency, the default value is 25
VMWI Ring Splash Len(sec)	Set the VMWI ring splash length, default is 0.5s.
Flash Time Max(sec)	Set the Max value of the device's flash time, the default value is 0.9
Flash Time Min(sec)	Set the Min value of the device's flash time, the default value is 0.1

Features and Call Forward

 Table 32 Features and call forward

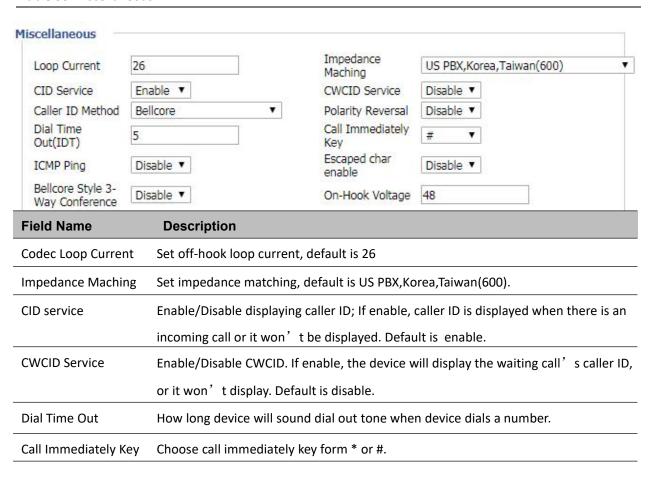


	Description
All Forward	Enable/Disable forward all calls
Busy Forward	Enable/Disable busy forward.
No Answer Forward	Enable/Disable no answer forward.
All Forward	Set the target phone number for all forward.
	The device will forward all calls to the phone number immediately
	when there is an incoming call.
Busy Forward	The phone number which the calls will be forwarded to when line
	is busy.
No Answer Forward	The phone number which the call will be forwarded to when
	there's no answer.
No Answer Timeout	The seconds to delay forwarding calls, if there is no answer at your
	phone.
Hold key code	Call hold signatures, default is *77.
Conference key code	Signature of the tripartite session, default is *88.
Transfer key code	Call forwarding signatures, default is *98.
IVR key code	Signatures of the voice menu, default is ****.
	Busy Forward No Answer Forward All Forward Busy Forward No Answer Forward No Answer Timeout Hold key code Conference key code Transfer key code

R key enable	Enable/Disable R key way call features.
R key cancel code	Set the R key cancel code, option are ranged from R1 to R9,
	default value is R1.
R key hold code	Set the R key hold code, options are ranged from R1 to R9, default
	value is R2.
R key transfer code	Set the R key transfer code, options are ranged from R1 to R9,
	default value is R4.
R key conference code	Set the R key conference code, options are ranged from R1 to R9,
	default value is R3.
R Key Reject 2nd Call	Set the R key Reject 2nd Call code, options are ranged from R1 to
Code	R9, default value is R0.
Speed Dial Code	Speed dial code, default is *74.

Miscellaneous

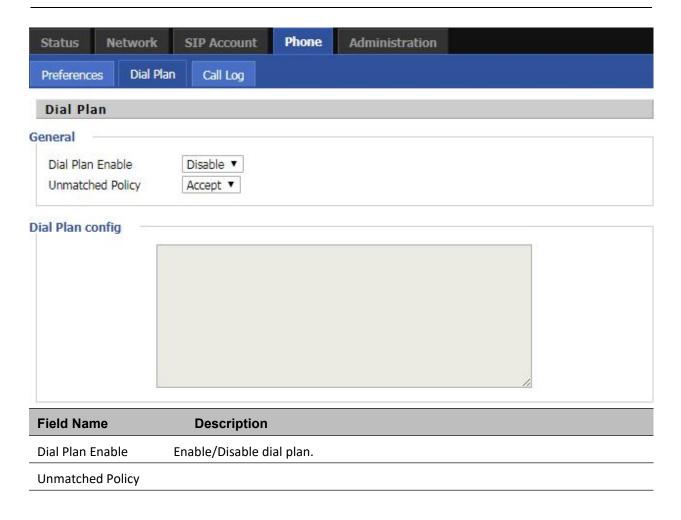
Table 33 Miscellaneous



ICMP Ping	Enable/Disable ICMP Ping.	
	If enable this option, home adapter will ping the SIP Server every interval	
	time, otherwise, It will send "hello" empty packet to the SIP Server.	
Escaped char enable	Open special character translation function; if enable, when you press the # key, it	
	will be translated to 23%, when disable, it is just #	

Dial Rule

Table 34 Dial Plan



Dial Plan Syntactic

Table 35 Dial Plan Syntactic

No.	String	Description
1	0123456789*#	Allowed characters
2	х	Lowercase letter x stands for one legal character

		To match one character form sequence. For example:
		[0-9]: match one digit form 0 to 9
3	[sequence]	[23-5*]: match one character from 2 or 3 or 4 or 5 or
		*
		x^0 x^1 x^2 x^3 x^n
4	x.	Match to , , ,
		For example:
		"01." :can match " 0" , "01" , "011" , " 0111" , "011" , "0111" , "01111"
5	<dialed:substituted></dialed:substituted>	Replace dialed with substituted. For example:
		<8:1650>123456: input is "85551212", output is "16505551212"
		Make outside dial tone after dialing "x", stop until dialing character "y"
		For example:
6	х,у	"9,1xxxxxxxxxx" : the device reports dial tone after inputting "9" , stops tone until inputting "1" $^{\prime\prime}$
		Set the delayed time. For example:
7	Т	"<9:111>T2": The device will dial out the matched number "111" after 2 seconds.

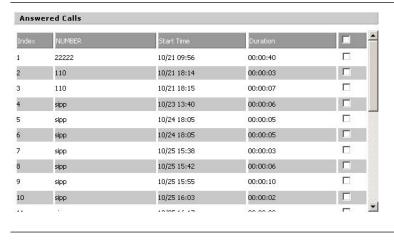
Call Log

To view the call \log information such as redial list , answered call and missed call

Table 36 Call log

Redial Calls Redial List Start Time 10/28 10:30 00:00:07 010123 10/28 12:02 00:00:01 П 010123 10/28 16:16 00:00:00 010123 10/28 16:16 00:00:00 123 10/28 16:20 00:00:13 123 Г 00:00:34 10/28 16:21 123 10/29 10:50 00:00:10 123 10/29 14:36 00:00:01 123 10/29 15:05 00:00:23 123 10/29 15:06 00:00:05

Answered Calls



Missed Calls



Administration

The user can manage the device in these webpages; you can configure the Time/Date, password, web access, system log and associated configuration TR069.

Management

Save config file

Table 37 Save Config File



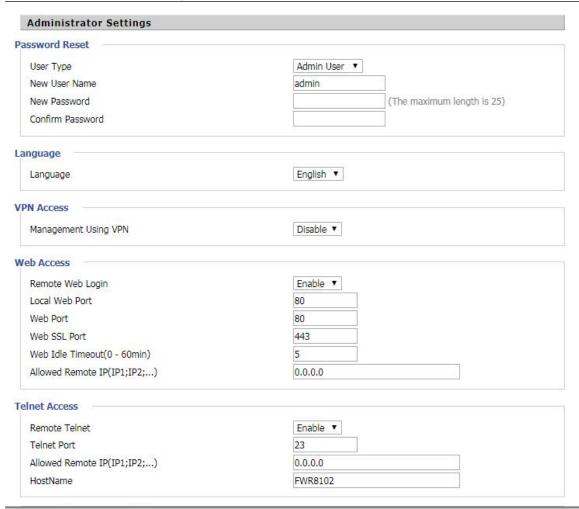
Field Name Description

Config file upload and Upload: click on browse, select file in the local, press the upload button to download begin uploading files

Download: click to download, and then select contains the path to download the configuration file

Administrator settings

Table 38 Administrator settings

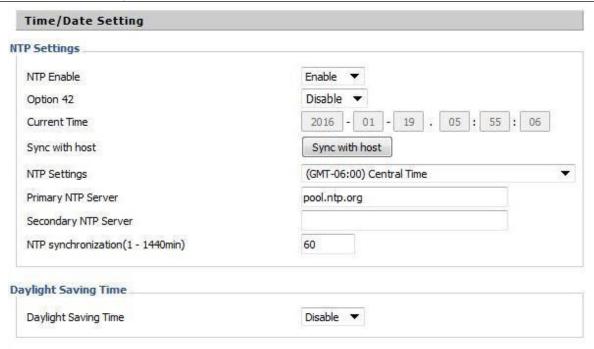


Field Name	Description
User type	Choose the user type from admin user and normal user and basic user
New User Name	You can modify the user name, set up a new user name
New Password	Input the new password
Confirm Password	Input the new password again
Language	Select the language for the web, the device support Chinese, English, and Spanish
	and so on
Remote Web Login	Enable/Disable remote Web login
Web Port	Set the port value which is used to login from Internet port and PC port, default is
	80

Web Idle timeout	Set the Web Idle timeout time. The webpage can be logged out after Web Idle
	Timeout without any operation
Allowed Remote	Set the IP from which a user can login the device remotely
IP(IP1,IP2,)	
Telnet Port	Set the port value which is used to telnet to the device

NTP settings

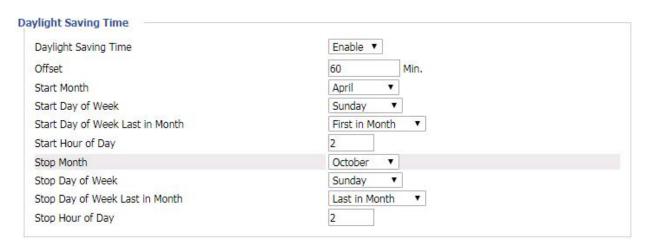
Table 39 NTP settings



Field Name	Description
NTP Enable	Enable/Disable NTP
Option 42	Enable/Disable DHCP option 42. This option specifies a list of the NTP servers
	available to the client by IP address
Current Time	Display current time
NTP Settings	Setting the Time Zone
Primary NTP Server	Primary NTP server's IP address or domain name
Secondary NTP Server	Options for NTP server's IP address or domain name
NTP synchronization	NTP synchronization cycle, cycle time can be 1 to 1440 minutes in any one, the
	default setting is 60 minutes

Daylight Saving Time

Table 40 Daylight Saving Time



Procedure

- Step 1. Enable Daylight Savings Time.
- Step 2. Set value of offset for Daylight Savings Time
- Step 3: Set starting Month/Week/Day/Hour in Start Month/Start Day of Week Last in Month/Start Day of Week/Start Hour of Day, analogously set stopping Month/Week/Day/Hour in Stop Month/Stop Day of Week Last in Month/Stop Day of Week/Stop Hour of Day.

Step 4.Press Saving button to save and press Reboot button to active changes.

System Log Setting

Table 41 System log Setting



Field Name	Description
Syslog Enable	Enable/Disable syslog function
Syslog Level	Select the system log, there is INFO and Debug two grades, the Debug INFO can
	provide more information
Remote Syslog Enable	Enable/Disable remote syslog function
Remote Syslog server	Add a remote server IP address
Syslog Enable	Enable/Disable syslog function

Factory Defaults Setting

Table 42 Factory Defaults Setting



Description

When enabled, the device may not be reset to factory defaults until this parameter is reset to Disable

Factory Defaults

Table 43 Factory Defaults

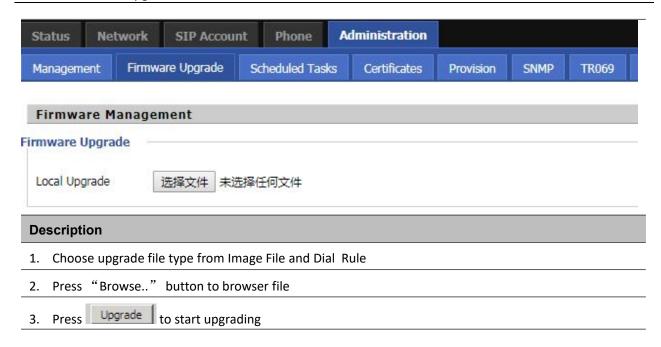


Description

Click Factory Default to restore the residential adapter to factory settings

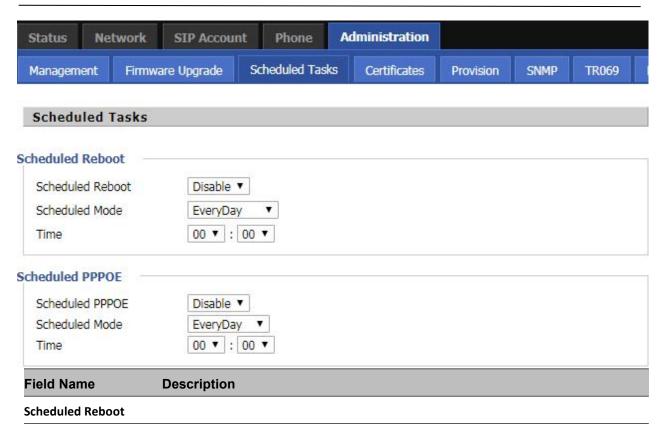
Firmware Upgrade

Table 44 Firmware upgrade



Scheduled Tasks

Table 45 Scheduled Tasks



Enable/Disable scheduled Reboot
Select scheduled Mode
Set the time to restart
Enable/Disable scheduled PPPoE
Select scheduled Mode
Set the time to start PPPoE

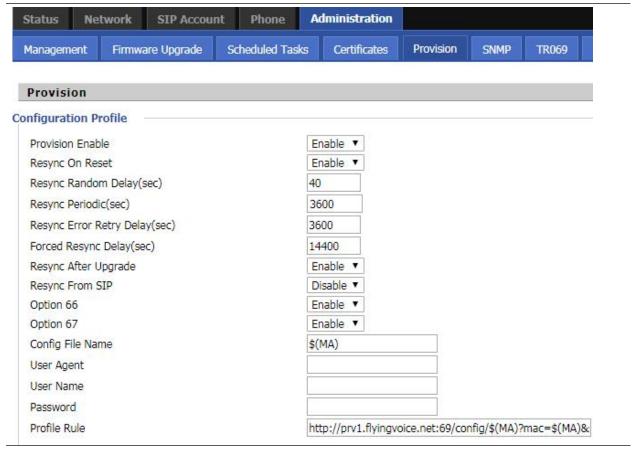
Provision

Provisioning allows the router to auto-upgrade and auto-configure devices which support TFTP, HTTP and HTTPs .

- Before testing or using TFTP, user should have tftp server and upgrading file and configuring file.
- Before testing or using HTTP, user should have http server and upgrading file and configuring file.
- Before testing or using HTTPS, user should have https server and upgrading file and configuring file
 and CA Certificate file (should same as https server's) and Client Certificate file and Private key file

User can upload a CA Certificate file and Client Certificate file and Private Key file in the Security page.

Table 46 Provision



Field Name	Description
Provision Enable	Enable provision or not.
Resync on Reset	Enable resync after restart or not
Resync Random	Set the maximum delay for the request of synchronization file. The default is 40
Resync Periodic(sec)	If the last resync was failure, The router will retry resync after the "Resync Error
Resync Error Retry	Set the periodic time for resync, default is 3600s
Forced Resync	If it's time to resync, but the device is busy now, in this case, the router will
Resync After	Enable firmware upgrade after resync or not. The default is Enabled
Resync From SIP	Enable/Disable resync from SIP
Option 66	It is used for In-house provision mode only. When use TFTP with option 66 to
Config File Name	It is used for In-house provision mode only. When use TFTP with option 66 to
Profile Rule	URL of profile provision file

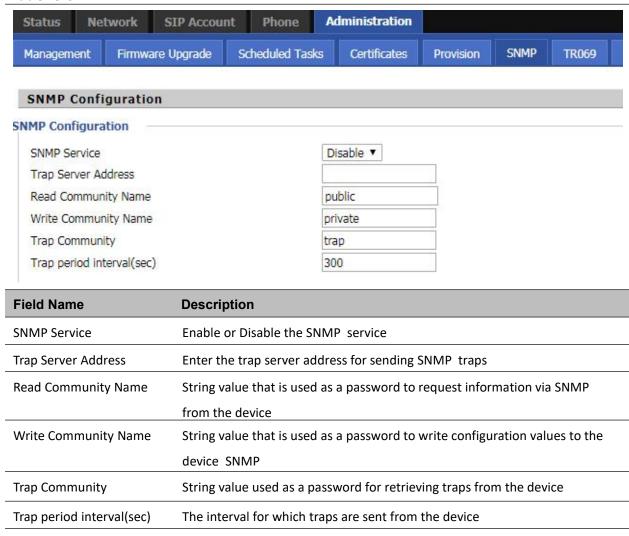
Table 47 Firmware Upgrade



Field Name	Description
Upgrade Enable	Enable firmware upgrade via provision or not
Upgrade Error Retry	If the last upgrade fails, the router will try upgrading
Delay(sec)	again after "Upgrade Error Retry Delay" period, default is 3600s
Upgrade Rule	URL of upgrade file

SNMP

Table 48 SNMP



TR-069

TR-069 provides the possibility of auto configuration of internet access devices and reduces the cost of management. TR-069 (short for Technical Report 069) is a DSL Forum technical specification entitled CPE WAN Management Protocol (CWMP). It defines an application layer protocol for remote management of end-user devices. Using TR-069, the terminals establish connection with the Auto Configuration Servers (ACS) and get configured automatically.

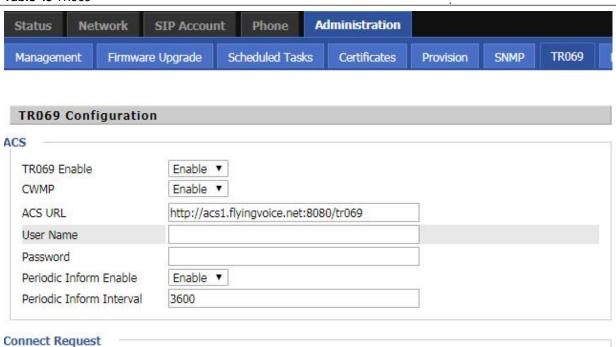
Device Configuration using TR-069

The TR-069 configuration page is available under Administration menu.

Table 49 TR069

User Name

Password



Field Name Description ACS parameters Enable or Disable TR069 TR069 Enable **CWMP Enable or Disable CWMP ACS URL** ACS URL address **User Name** ACS username Password ACS password Periodic Inform Enable the function of periodic inform or not. By default it is Enabled Periodic Inform Periodic notification interval with the unit in seconds. The default value is 3600s **Connect Request parameters User Name** The username used to connect the TR069 server to the DUT. The password used to connect the TR069 server to the DUT. Password

FGW4148-16S

Diagnosis

In this page, user can do packet trace, ping test and traceroute test to diagnose the device's connection status.

Table 50 Diagnosis

Description

1. Packet Trace

Users can use the packet trace feature to intercept packets which traverse the device. Click the Start button to start home adapter tracking and keep refreshing the page until the message trace shows to stop, click the Save button to save captured packets.

2. Ping Test

Enter the destination IP or host name, and then click Apply, device will perform ping test.

Ping Test		
Dest IP/Host Name		
WAN Interface	1_TR069_VOICE_INTERNET_R_VID_ 🔻	
PING www.baidu.c	om (115.239.210.26): 56 data bytes	•
64 bytes from 115	.239.210.26: seq=0 ttl=54 time=43.979 ms	9
64 bytes from 115	.239.210.26: seq=1 ttl=54 time=53.875 ms	
64 bytes from 115	.239.210.26: seq=2 ttl=54 time=45.226 ms	
64 bytes from 115	.239.210.26: seq=3 ttl=54 time=49.534 ms	
64 bytes from 115	.239.210.26: seq=4 ttl=54 time=49.045 ms	
www.baidu.con	n ping statistics	
- 1 - 1	ted, 5 packets received, 0% packet loss	¥
5 packets transmit		

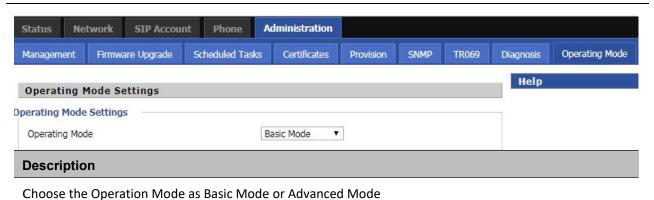
3. Traceroute Test

Enter the destination IP or host name, and then click Apply, device will perform traceroute test.



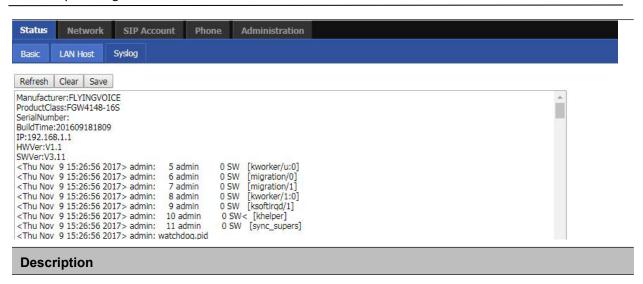
Operating Mode

Table 51 Operating mode



System Log

Table 52 System log



If you enable the system log in Status/syslog webpage, you can view the system log in this webpage.

Logout

Table 53 Logout



Press the logout button to logout, and then the login window will appear.

Reboot

Press the Reboot button to reboot the device.

Chapter 5 IPv6 address configuration

The router devices support IPv6 addressing. This chapter covers:

- Introduction
- IPv6 Advance
- Configuring IPv6
- Viewing WAN port status
- IPv6 DHCP configuration for LAN/WLAN clients
- LAN DHCPv6

Introduction

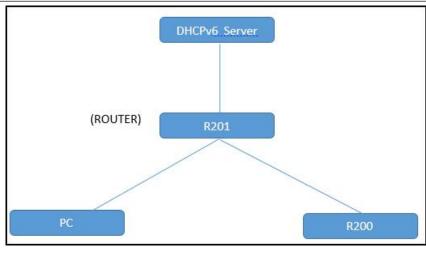
DHCPv6 protocol is used to automatically provision/configure IPv6 capable end points in a local network. In addition to acquiring an IPv6 IP address for the WAN interface and its associated LAN/WLAN clients, the devices are also capable of prefix delegation.

The Routers devices support the following types of modes of IPv6 addresses:

- Stateless DHCPv6
- Statefull DHCPv6

Table 54 IPv6 Modes

Mode	Description
Stateless	In Stateless DHCPv6 mode, the Routers devices listen for ICMPv6 Router
	Advertisements messages which are periodically sent out by the routers on the
	local link or requested by the node using a Router Advertisements solicitation
	message. The device derives a unique IPv6 address using prefix receives from the
	router and its own MAC address.



Statefull

In Statefull DHCPv6 mode, the client works exactly as IPv4 DHCP, in which hosts receive both their IPv6 addresses and additional parameters from the DHCP server.

IPv6 Advance

To enable IPv6 functionality:

Navigate to Network > IPv6 Advanced page.

Select Enable from the IPv6 Enable drop-down list.

Click Save.

Table 55 Enabling IPv6



Type commands, here PC can get the default adapter: fe80::221:f2ff:fe02:1a4f%15.

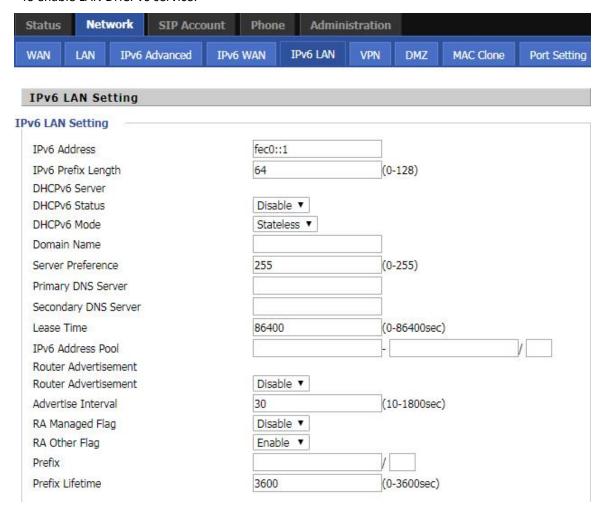
```
FØ-DE-F1-C8-96-66
DHCP 已启用。
自动配置已启用
                                   是是
  地链接 IPv6 地址
                                   fe80::2db3:666d:88d9:d1c2x15(首选)
                                   192.168.11.90(首选)
                                   255.255.255.0
    粗约的时间
                                   2016年2月2日 星期二
2016年2月3日 星期三
                                                        上午 10:43:34
                                                      上牛 10:43:33
    过期的时间
                                   fe80::221:f2ff:fe02:1a4fx15
                                   192.168.11.1
DHCP 服务器
                                   192.168.11.1
DHCPv6 IAID
                                   368107249
DHCPv6 客户端 DUID
                                   00-01-00-01-1E-0A-7B-28-F0-DE-F1-C8-96-66
DNS 服务器
                                   192.168.11.1
                                   192.168.10.1
TCPIP 上的 NetBIOS
```

We can ping through this address.

LAN DHCPv6

When IPv6 is enabled, the LAN/WLAN clients of Routers can be configured to receive IPv6 addresses from locally configured IPv6 pool or from an external DHCPv6 server.

To enable LAN DHCPv6 service:

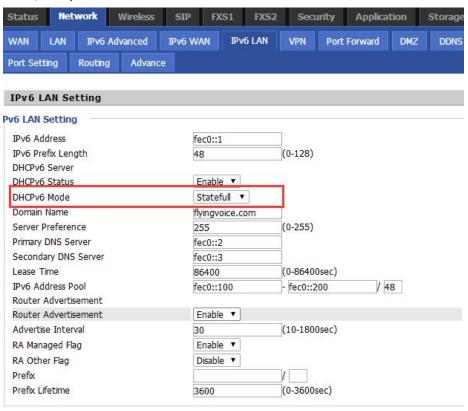


Configuring IPv6

Configuring Statefull IPv6

1. Navigate to Network > IPv6WAN page. The following window is displayed:

Stateless mode dhcpv6 client send ipv6 address and DNS request, and the server reply the DNS server and Domain name, and ipv6 address.



DHCPv6 client configure as this

Table 56 Configuring Statefull IPv6



In this way the Router B can get ipv6 address and DNS address, for more information please check the packets dhcpv6 stateful.pcap.

Also we can check via CLI.

```
Link encap:Ethernet HWaddr 00:21:F2:08:16:59
inet addr:192.168.11.64 Bcast:192.168.11.255 Mask:255.255.255.0
inet6 addr: fec0::100/128 Scope:Site
inet6 addr: fe80::221:f2ff:fe08:1659/64 Scope:Link
UP BROADCAST RUNNING MULTICAST MTU:1500 Metric:1
RX packets:1235 errors:0 dropped:0 overruns:0 frame:0
TX packets:1346 errors:0 dropped:0 overruns:0 carrier:0
collisions:0 txqueuelen:0
RX bytes:103756 (101.3 KiB) TX bytes:1329724 (1.2 MiB)
```

In this mode ,PC can get ipv6 address too.

```
" tat /etc/resolv.conf
nameserver 192.168.11.1
nameserver 192.168.10.1
nameserver fec0::2
nameserver fec0::3
```

```
以太网适配器 本地连接:
       詩定的 DNS 后缀 . . .
                                    : flyingvoice.com
                                    : Intel(R) 82579LM Gigabit Network Conne
  物理地址-
                                      FØ-DE-F1-C8-96-66
  物理地址.....
DHCP 已启用 .....
自动配置已启用....
本地站点的 IPv6 地址.
获得租约的时间 ...
                                    :fec0::101x1(首选)
                                                          上午 11:47:06
上午 11:47:06
                                    : 2016年2月2日
     的过期的时间
                                      2016年2月3日
                                      fe80::2db3:666d:88d9:d1c2x15(首选)
           IPv6 地址.
  IPv4 地址
                                      192.168.11.90(首选)
   上网掩码
                                      255.255.255.0
                                      2016年2月2日
      粗约的时间
                                                            上午 11:02:22
     的过期的时间
                                      2016年2月3日
                                      fe80::221:f2ff:fe02:1a4fx15
                                      192.168.11.1
  DHCP 服务器
                                    : 192.168.11.1
  DHCPv6 IAID
                                    : 368107249
  DHCPv6 客户端 DUID
                                    : 00-01-00-01-1E-0A-7B-28-F0-DE-F1-C8-96
                              . . . : fec0::2x1
                                      192.168.11.1
```

Configuring Stateless IPv6

Stateless mode dhcpv6 client only send DNS request, and the server reply the DNS server and Domain name, DHCPv6 server configure as the picture shows, DHCPv6 client configure as this.



After the configuration, we can check the packets about the dhcpv6 client. For more inforation, please check dhcpv6_stateless.pcap.

In the router we can check via CLI.But you can not see ipv6 address

```
# cat /etc/resolv.conf
nameserver 192.168.11.1
nameserver 192.168.10.1
nameserver fec0::2
aameserver fec0::3
```

ping DNS address

When in stateful mode, device can get ipv6 address from dhcpv6 server, in this way we can ping a dns, verify if it can do domain name resolve by ipv6 dns address. Check this we can use the packets to prove. You can find on this packet dns.pcap

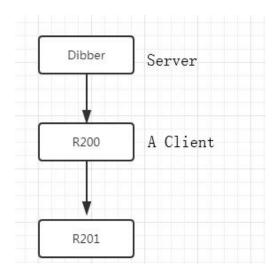
WAN DHCPv6 client

As I show you with router B, router B use the WAN DHCPv6 client, you can refer to the part 2 to check the stateful and stateless mode. Another thing I need show you is Prefix distribution.

Prefix distribution

- 1)WAN port DHCPv6 server enable prefix distribution feature.
- 2)WAN port DHCPv6 client enable prefix distribution
- 3)LAN port DHCPv6 server disable dhcp service

Topo like this



Dibber dhcpv6 server configuration see the attach.

This time I use the stateful mode Router A can get an ipv6 address ,WAN port and LAN port both can get ipv6 address from the dibbler server.

```
br0 Link encap:Ethernet HWaddr 00:21:F2:02:1A:4F
inet addr:192.168.11.1 Bcast:192.168.11.255 Mask:255.255.255.0
inet6 addr: 2001:db8:352e:0:221:f2ff:fe02:la4f/48 Scope:Global
inet6 addr: fe80::221:f2ff:fe02:la4f/64 Scope:Link
inet6 addr: fec0::1/48 Scope:Site
UP BROADCAST RUNNING MULTICAST MTU:1500 Metric:1
RX packets:1255 errors:0 dropped:0 overruns:0 frame:0
TX packets:1550 errors:0 dropped:0 overruns:0 carrier:0
collisions:0 txqueuelen:0
RX bytes:226565 (221.2 KiB) TX bytes:613327 (598.9 KiB)
```

The behind router also get an ipv6 address

```
eth2.1 Link encap:Ethernet HWaddr 00:21:F2:08:16:59
inet addr:192.168.11.64 Bcast:192.168.11.255 Mask:255.255.255.0
inet6 addr: 2001:db8:352e:0:221:f2ff:fe08:1659/64 Scope:Global
inet6 addr: fe80::221:f2ff:fe08:1659/64 Scope:Link
UP BROADCAST RUNNING MULTICAST MTU:1500 Metric:1
RX packets:693 errors:0 dropped:0 overruns:0 frame:0
TX packets:563 errors:0 dropped:0 overruns:0 carrier:0
collisions:0 txqueuelen:0
RX bytes:58337 (56.9 KiB) TX bytes:375603 (366.7 KiB)
```

Use this ipv6 address we can ping Router A LAN ipv6 address

```
C: Wsers Administrator>ping 2001:db8:352e:0:221:f2ff:fe02:1a4f
正在 Ping 2001:db8:352e:0:221:f2ff:fe02:1a4f 具有 32 字节的数据:
来自 2001:db8:352e:0:221:f2ff:fe02:1a4f 的回复: 时间<1ms
来自 2001:db8:352e:0:221:f2ff:fe02:1a4f 的回复: 时间<1ms
来自 2001:db8:352e:0:221:f2ff:fe02:1a4f 的回复: 时间<1ms
来自 2001:db8:352e:0:221:f2ff:fe02:1a4f 的回复: 时间<1ms
```

Viewing WAN/LAN port status

To view the status of WAN port:

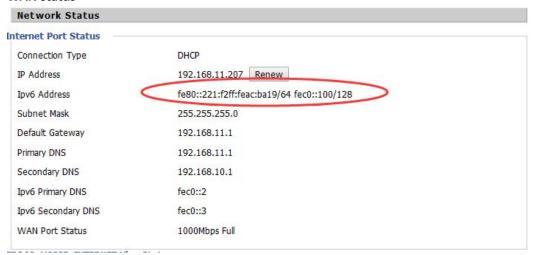
Navigate to Status page.

LAN status

Now only LAN ipv6 dhcpv6 mode is statefull, it can display ipv6 client ipv6 address



WAN status



Chapter 6 Troubleshooting Guide

This chapter covers:

- Configuring PC to get IP Address automatically
- Cannot connect to the Web
- Forgotten Password

Configuring PC to get IP Address automatically

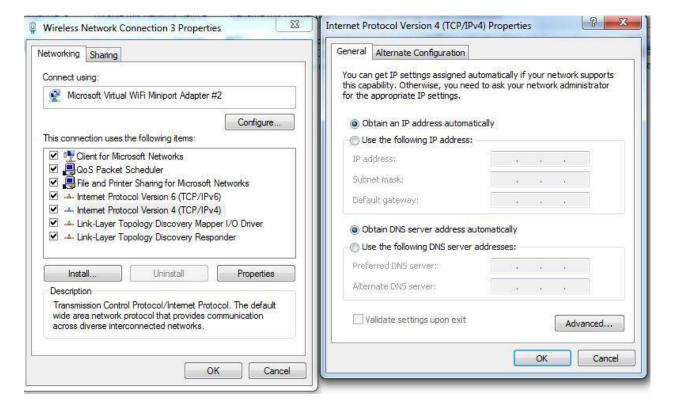
Follow the below process to set your PC to get an IP address automatically:

Step 1 : Click the "Start" button

Step 2: Select "control panel", then double click "network connections" in the "control panel"

Step 3: Right click the "network connection" that your PC uses, select "attribute" and you can see the interface as shown in Figure 3.

Step 4.: Select "Internet Protocol (TCP/IP)", click "attribute" button, then click the "Get IP address automatically".



Cannot connect to the Web

Solution:

- Check if the Ethernet cable is properly connected
- Check if the URL is correct. The format of URL is: http:// the IP address
- Check on any other browser apart from Internet explorer such Google
- Contact your administrator, supplier or ITSP for more information or assistance.

Forgotten Password

If you have forgotten the management password, you cannot access the configuration web GUI. Solution:

To factory default: press and hold reset button for 10 seconds.