



TA400/TA800

User Manual

Sales Tel: +86-592-5503309
E-mail: sales@yeastar.com
Support Tel: +86-592-5503301
E-mail: support@yeastar.com
Web: <http://www.yeastar.com>

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WEEE Warning



In accordance with the requirements of council directive 2002/96/EC on Waste of Electrical and Electronic Equipment (WEEE), ensure that at end-of-life you separate this product from other waste and scrap and deliver to the WEEE collection system in your country for recycling.

Contents

| | |
|---|-----------|
| About This Guide | 5 |
| Getting Started | 6 |
| Accessing Web GUI..... | 6 |
| Web Configuration Panel..... | 7 |
| Making and Receiving Calls | 7 |
| Basic Operations | 8 |
| Voice Menu | 8 |
| Call Hold | 9 |
| Call Waiting..... | 9 |
| Call Transfer..... | 9 |
| Three-party Conference..... | 10 |
| Direct IP Call | 10 |
| Change TA400/800's IP Address Using Analog Phones | 10 |
| FXS Port Settings | 12 |
| General Settings | 12 |
| Other Settings | 14 |
| Hunt Group | 15 |
| VoIP Settings | 18 |
| VoIP Server Settings..... | 18 |
| Dial Pattern Template..... | 20 |
| SIP Settings | 21 |
| IAX Settings | 26 |
| Gateway Settings | 28 |
| Basic Preferences..... | 28 |
| Feature Codes | 29 |
| Speed Dial..... | 31 |
| Prompt Settings | 32 |
| Custom Prompts | 32 |
| Music on Hold Prompts..... | 33 |
| System Prompts..... | 33 |
| Advanced Settings | 35 |
| Distinctive Ringtones | 35 |
| Tone Zone Settings | 35 |
| RADIUS Settings | 36 |

| | |
|----------------------------------|-----------|
| Network Preferences | 38 |
| LAN Settings | 38 |
| Service | 39 |
| VLAN Settings..... | 40 |
| VPN Settings..... | 40 |
| DDNS Settings..... | 41 |
| Static Route..... | 42 |
| TR-069 Settings..... | 43 |
| Security Center | 45 |
| Security Center | 45 |
| Alert Settings..... | 46 |
| AMI Settings..... | 48 |
| Certificates | 49 |
| Firewall Rules | 50 |
| IP Blacklist | 52 |
| System Preferences | 54 |
| Password Settings | 54 |
| Date and Time..... | 54 |
| Auto Provision Settings..... | 55 |
| Firmware Update | 57 |
| Upgrade through HTTP | 57 |
| Upgrade through TFTP | 58 |
| Backup and Restore | 59 |
| Reset and Reboot..... | 59 |
| Status | 61 |
| FXS Port Status | 61 |
| Network status | 62 |
| System Info..... | 62 |
| Reports | 63 |
| Call Logs | 63 |
| System Logs | 63 |
| Packet Tool | 64 |
| Port Monitor Tool..... | 64 |

About This Guide

Yeastar TA400/800 FXS Analog VoIP Gateways are cutting-edge products that connect legacy telephones, fax machines and PBX systems with IP telephony networks and IP-based PBX systems. Featuring rich functionalities and easy configuration, Yeastar TA is ideal for small and medium enterprises that wish to integrate a traditional phone system into IP-based system. Yeastar TA helps them to preserve previous investment on legacy telephone system and reduce communication costs significantly with the true benefits of VoIP.

Audience

This manual will help you learn how to operate and manage your TA400/800 FXS Analog VoIP Gateway. In this guide, we describe every detail on the functionality and configuration of TA400/800. We begin by assuming that you are interested in TA400/800 and familiar with networking and other IT disciplines.

Safety when working with electricity



- Do not open the device when the device is powered on.
- Do not work on the device, connect or disconnect cables when lightning strikes.

Features Highlights

- 4/8 FXS ports
- Fully compliant with SIP and IAX2
- Hunt Group
- Configurable VoIP Server templates
- Reliable FAX performance with T.38
- 3-party conference
- Direct IP Calling
- Blind/Attended Transfer
- Support RADIUS protocol

Check the TA400/800 Installation Guide here:

http://www.yeastar.com/downloadFile/Yeastar_TA_Series_Installation_Guide_en.pdf

For more information, please click:

<http://www.yeastar.com/Products.html/Analog-VoIP-Gateways>

Getting Started

In this chapter, we guide you through the basic steps to start with a new TA400/800:

- [Accessing Web GUI](#)
- [Web Configuration Panel](#)
- [Making and Receiving Calls](#)

Accessing Web GUI

The TA400/800 attempts to contact a DHCP server in your network to obtain valid network settings (e.g., the IP address, subnet mask, default gateway address and DNS address) by default.

Please enable DHCP Server in your network to obtain the TA400/800 IP address.

How to check TA400/800 IP address:

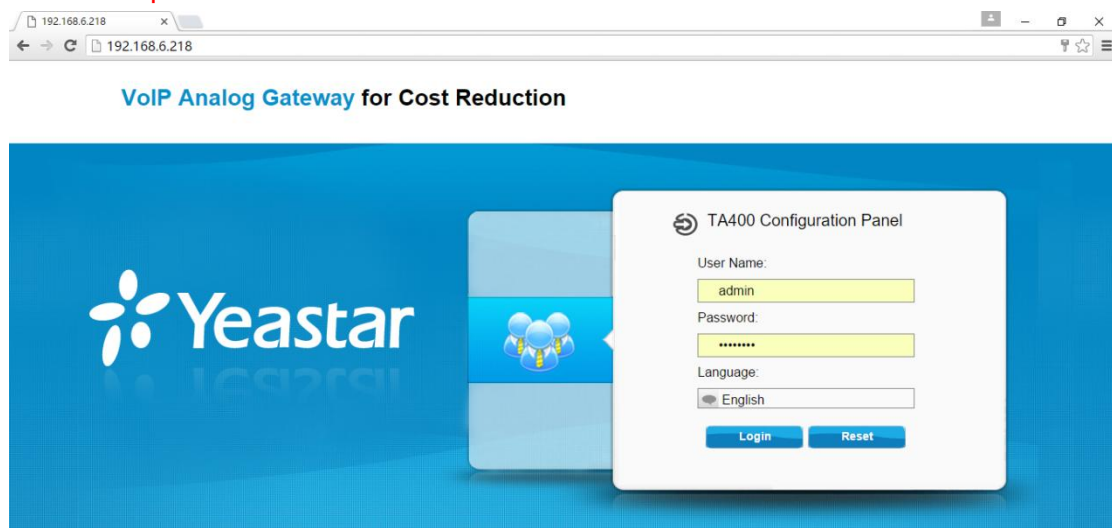
1. Pick up the analog phone, then access the voice menu prompt by dialing “***”.
2. Dial “1” to check the IP address.
3. Dial “2” for web access address.

After entering the IP address in the web browser, users will see a log-in screen.

Check the default settings below:

Username: **admin**

Password: **password**



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Figure 2-1 TA400/800 Login page

Web Configuration Panel

There are 4 main sections on the Web Configuration Panel for users to check the TA400/800's status and configure it.

- **Status:** check System Status, Port Status, Trunk Status, Network Status and check call logs, system logs.
- **System:** configure Network Settings, Security related Settings, System Date and Time, Password, Backup and Restore, etc.
- **Gateway:** configure FXS ports, gateway settings and SIP settings, etc.
- **Logout:** log out TA400/800.

Note:

After saving the changes, remember to click the “Apply changes” button on the upper right corner of the Web GUI to make the changes take effect.

Making and Receiving Calls

You will need an active VoIP account from a VoIP service provider or PBX, which provides a VoIP telephone number to allow you to make and receive calls.

- **Making internal calls**
To place an internal call, pick up the analog phone and dial the other port's caller ID number or dial *99 followed by the port number.
For example, to reach the other port with caller ID number 300, dial “300”.
To reach port 2, you can dial *9902 to ring the phone connected to port 2.
- **Making outbound calls**
To make an outbound call, you need to dial according to the FXS port dial pattern.
By the default dial pattern, you can dial the desired outgoing number directly.
- **Answering calls**
To answer a call, pick up the handset as you usually do.

Basic Operations

In this chapter, we give instructions about how to operate on analog phones connected to TA400/800 to use some features.

- Voice Menu
- Call Hold
- Call Waiting
- Call Transfer
- Three-party Conference
- Direct IP Call
- Change TA400/800's IP Address

Voice Menu

TA400/800 provides a voice menu to guide you to configure the network settings for the device. You need to press *** on the analog phone which is connected to TA400/800's FXS port to enter the voice menu.

The default password to enter "Advanced Settings" is 123456. You can change the password on TA400/800 Web page.(Gateway→Gateway Settings→Feature Codes→Voice Menu Password Settings)

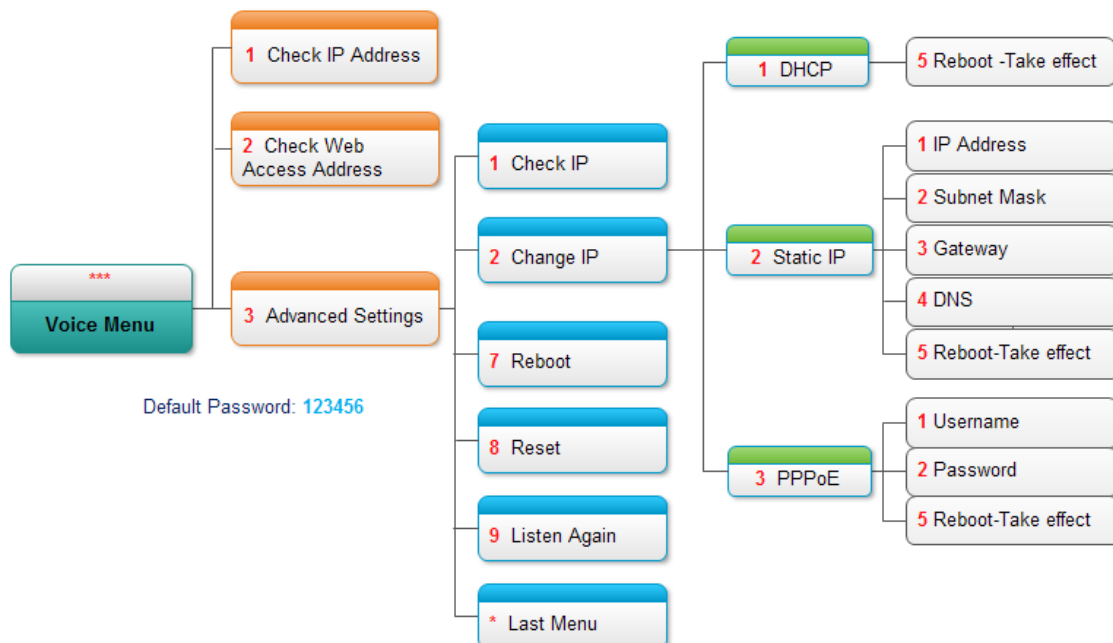


Figure 3-1 TA400/800 Voice Menu

Tips:

1. Press "9" to listen to the prompt again.
2. Press "*" to return to the last menu.

Call Hold

An active call can be held by pressing "flash" key on the analog phone. Press the key again to resume the call.

If there is no "flash" key on the phone, you can use "hook flash" (quickly toggle on-off hook) to hold a call. The call may be disconnected by chance if using "hook flash".

Call Waiting

If the call waiting is activated for the FXS port, the FXS user who is in a call can hear a call waiting tone "beep" when there is a new incoming call. The user can press "hook flash" to toggle between the active call and the incoming call.

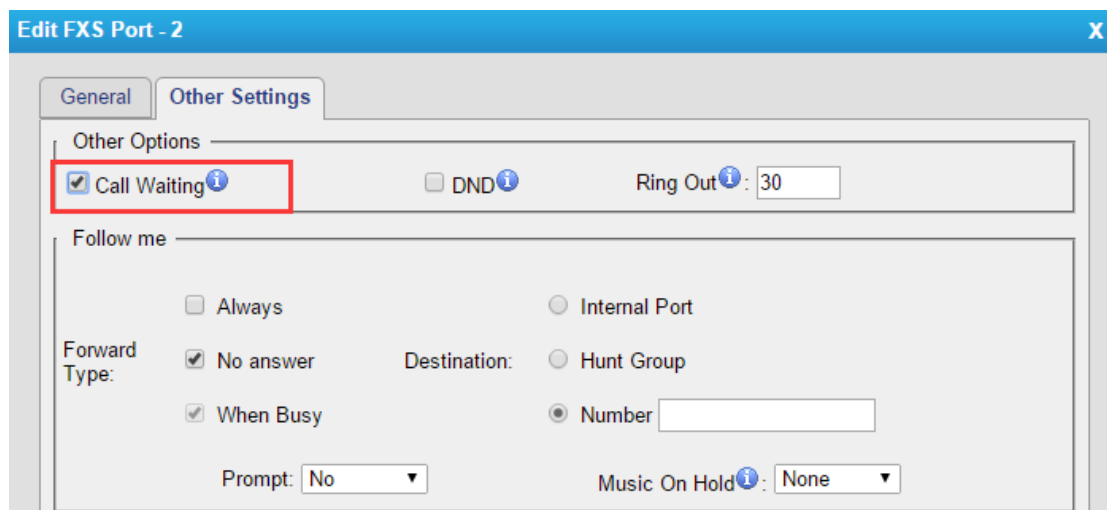


Figure 3-2 Enable Call Waiting

Call Transfer

Blind transfer and attended transfer are supported on TA400/800. Users can achieve call transfer by pressing the feature code during the call.

Blind Transfer

Default feature code: *03

1. Dial "*03" during the call;
2. Dial the called number after hearing a prompt "transfer";
3. The call will be transferred after the number is dialed.

Attended Transfer

Default feature code: *3

1. Dial "*3" during the call;
2. Dial the called number after hearing a prompt "transfer";
3. Talk to the transfer recipient;
4. The call will be transferred after hanging up.

Three-party Conference

Users can make a three-party conference call on TA400/800.

Assuming that A and B are in the call and B wants to invite C to a conference. Please check the following steps of how to establish a conference.

1. B presses "flash" key or taps hook flash to get a dial tone; A will hear the on hold music meanwhile;
2. B dials C's number;
3. If C answers the call, then B presses "flash" key or tap hook flash, the conference will be established, including A, B, and C.
4. If there is no answer on C, A can press "flash" key or tap hook flash to resume the call with A.
5. C will be ejected if B presses "flash" key or taps flash hook during the conference call.

Direct IP Call

Direct IP call allows two parties, that is, a FXS Port with an analog phone and another VoIP Device, to talk to each other in an ad hoc fashion without a SIP proxy. The default "Direct IP Calling" feature code is *96.

Example:

Target IP address: 192.168.2.123

Destination port: 5060

To call the IP phone, you should dial *96192*168*2*123*5060 on the analog phone.

Change TA400/800's IP Address Using Analog Phones

By default, TA400/800 obtains a dynamic IP address from the DHCP server. You can change the device's IP address via the analog phone which is connected to the FXS port.

There are 3 modes supported on TA400/800 access the internet.

- DHCP
- Static IP Address
- PPPoE

Here we introduce how to set a static IP address for TA400/800.

IP address: 192.168.10.125

Subnet mask: 255.255.255.0

Gateway: 192.168.10.1


DNS: 8.8.8.8

1. Press *** to enter the voice menu.
2. Press 3 to enter the "Advanced Settings".
3. Enter the password follow by the pound key: 123456# (The default password is 123456).
4. Press 2 to change the IP address.
5. Press 2 to enable the static IP.
6. Press 1 to change the IP address and follow by the new IP address (1192*168*10*125).
7. Press 2 to change the subnet mask and follow by the new subnet mask (2255*255*255*0).
8. Press 3 to change the gateway and follow by the new gateway (3192*168*10*1).
9. Press 4 to change the DNS and followed by the new DNS (48*8*8*8).
10. Press 5 to reboot the device.
11. After reboot, you can access the device by the new IP address.

FXS Port Settings

This chapter explains how to configure FXS port on TA400/800, go to **Gateway**→**Port List**→**Port List** page to configure the FXS ports.

- [General Settings](#)
- [Other Settings](#)
- [Hunt Group](#)

Click "Edit" button  to configure the FXS port.

General Settings

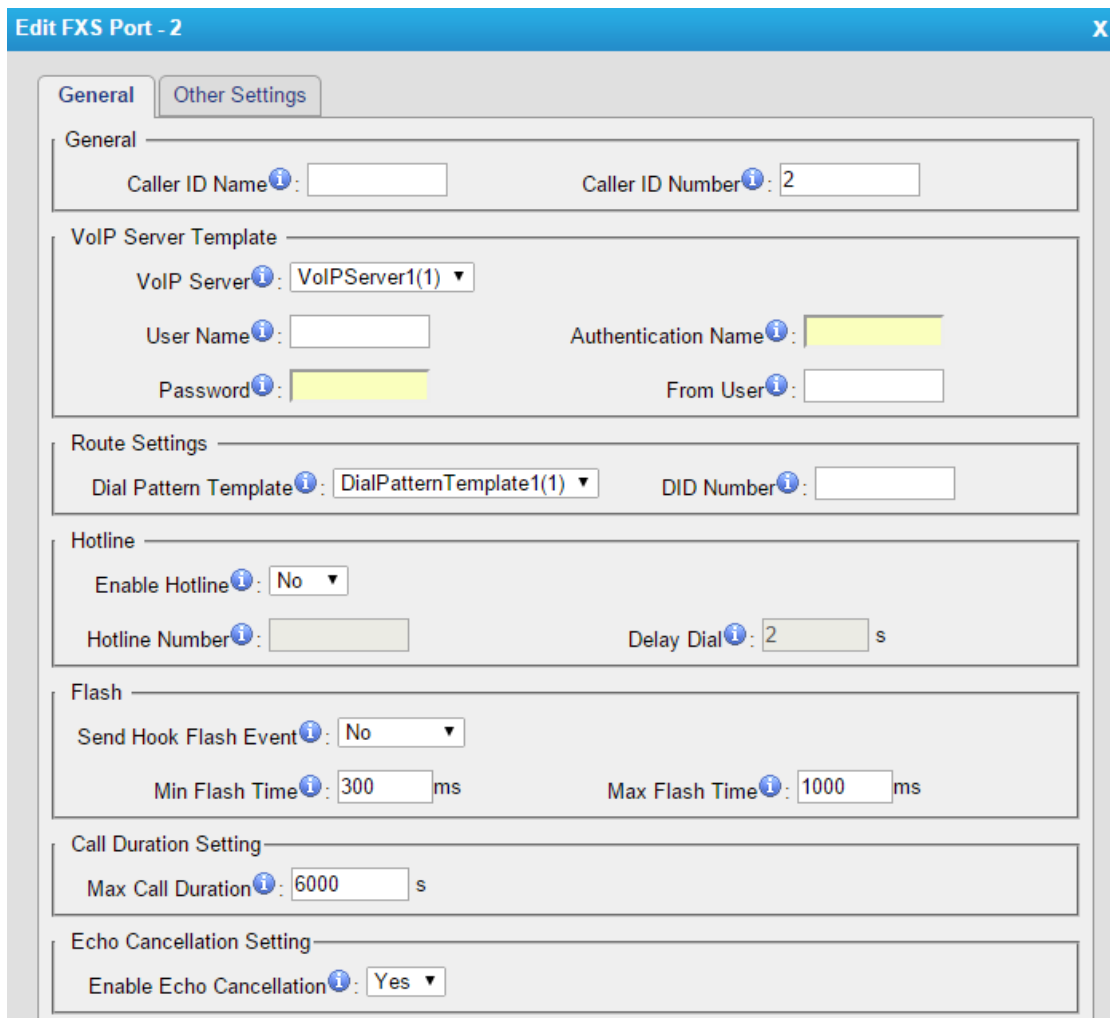


Figure 4-1 FXS Port General Settings

Table 4-1 Description of FXS Port General Settings

| Items | Description | |
|----------------|----------------|---|
| General | Caller ID Name | A character-based name for this user. For |

| | | |
|----------------------------------|-----------------------|---|
| | | example, Bob Jones. |
| | Caller ID Number | The Caller ID string used when this user calls another user. |
| VoIP Server Template | VoIP Server | Choose a VoIP server, where the account will be registered. |
| | User Name | User name provided by VoIP server. |
| | Authentication Name | Authentication name provided by VoIP server. |
| | Password | Authentication password provided by VoIP server. |
| | From User | Provided by VoIP server. Keep this field blank if not needed. |
| Route Settings | Dial Pattern Template | The account will be allowed to make outbound calls according to the selected template. |
| | DID Number | Defines the expected DID number if this trunk passes DID on incoming calls. Leave this field blank to match calls with any or no DID info. |
| Hotline | Enable Hotline | Select whether to use Hotline or not. Hotline is disabled by default. If this feature is enabled, the system will dial out the hotline number automatically after off-hook. |
| | Hotline Number | Set the number to dial out automatically after off-hook. |
| | Delay Dial | Define how long to make Hotline take effect after you pick up the phone. |
| Flash | Send Hook Flash Event | If the option "Send Hook Flash Event" is set to "Yes", when pressing the flash key on the analog phone during an active call. TA will send a SIP DTMF flash event to the VoIP server. It can be an interoperation with a FXO gateway which is used for informing FXO gateway to generate the flash signal to the CO line. |
| | Min Flash Time | Sets the minimum time the phone is on hook before being detected as a hook flash. |
| | Max Flash Time | Sets the maximum flash time. |
| Call Duration Setting | | Setup the max call duration for every call of this user, but it's only valid for outbound calls. Note: this setting will not be valid for internal calls. |
| Echo Cancellation Setting | | To disable or enable echo cancellation. If the line is used to connect POS machine, it is suggested that echo cancellation is disabled. |

Other Settings

Figure 4-2 FXS Port Other Settings

Table 4-2 Description of FXS Port Other Settings

| Follow Me | |
|--|---|
| <p>Choose the forward type and configure the relevant destination number. In different conditions, the incoming calls to the account will be forwarded to different destinations.</p> <ul style="list-style-type: none"> • Always: always forward calls to the destination number. • No Answer: forward calls when no one answers the call. • When Busy: forward the call when the account is busy. | |
| Features | |
| Call Waiting | Check this option if the extension should have Call Waiting capability. If this option is checked, the “When busy” follow me options will not be available. |
| DND | Don’t Disturb. When DND is enabled for an extension, the extension will not be available. |

| | |
|---|---|
| Ring Out | Check this option if you want to customize the ring time. Ring tone will stop over the time defined. |
| Volume Settings | |
| Rxgain | Adjust receive gain. |
| Txgain | Adjust transmit gain. |
| Caller ID Settings | |
| Caller ID Signalling | choose the Caller ID signaling: <ul style="list-style-type: none"> • FSK • DTMF. |
| Sending Mode | choose the Caller ID Sending mode: <ul style="list-style-type: none"> • Ring + Caller ID + Ring • Caller ID + Ring • Polarity + Caller ID + Ring <p>Note: if the FXS port is connected to a device's FXO port, Caller ID settings on FXS port and FXO port should be the same, or the call cannot be established.</p> |
| Caller ID Type | choose the Caller ID Type: <ul style="list-style-type: none"> • Bell-USA • V23-UK • V23-NTT Japan |
| Fax | |
| Enable T.38 | If the FXS port is connected to a Fax machine, this option should be checked. |
| MWI Settings | |
| Subscribe for MWI | Message Waiting Indicator (MWI) refers to a special LED on the analog phone, which indicates the voice mail status. |
| MWI Light Option | There are three lighting methods supported on TA FXS Gateway: neon, FSK and Polarity. Choose the lighting method according to your phone. TA FXS Gateway will light the lamp up once receiving a new message; the lamp will go off if there is no unread voice mail. |
| Polarity Settings | |
| A polarity signal is required on some systems for reverse polarity billing. Select if reverse polarity signal is activated on this FXS port. It is not activated by default. <ul style="list-style-type: none"> • Polarity Answer: Whether to send polarity signal while picking up the phone. • Polarity Hangup: Whether to send polarity signal while hanging up the phone. | |

Hunt Group

Hunt group is a feature that allows a call reaching multiple FXS ports. The FXS ports will act as a single group, called a hunt group. The number of hunt groups is limited by the number of ports each TA FXS Gateway model has. For example, there are 24 hunt

groups on TA FXS Gateway2400.

Hunt group will be chosen when configuring the FXS port "Follow Me". The hunt group will work when a call reaches the FXS port associated user which is busy or no answer.

Figure 4-3 Hunt Group

Figure 4-4 Set Hount Group on "FXS Port" Page

There are 3 strategies for hunt group on TA FXS Gateway.

- **Simultaneous**
All the FXS users will ring at the same time.
- **Circular**
In circular hunting, the calls are processed "round-robin". If a call is delivered to FXS port1, the next call will go to 2, the next to 3. The succession throughout each

of the FXS users even if one of the previous local users becomes free. When the end of the hunting group is reached, the hunting starts over at the first local users.

- **Linear**

Linear hunting is also referred to serial hunting. In linear hunting, calls are always delivered to the first FXS users, unless it is busy, then the second, third, and so on.

VoIP Settings

To integrate with other IPPBX, we need to configure the VoIP settings in TA FXS Gateway to set up VoIP trunk (SIP and IAX). In this chapter, we introduce the following settings:

- [VoIP Server Settings](#)
- [Dial Pattern Template](#)
- [SIP Settings](#)
- [IAX Settings](#)

VoIP Server Settings

There are some configurable VoIP (SIP/IAX) Server templates on this page. The number of VoIP Server templates is the half of FXS ports on TA FXS Gateway. The VoIP server settings help the FXS ports to register to the VoIP server. Once configured, the templates can be chosen on FXS port setting page.

There are 3 register modes for VoIP server. Users could select one mode for the VoIP server and apply it to FXS ports.

- **Service Provider** - IP Based VoIP Provider, do not generally require the TA gateway to register with the provider. Only IP address or domain is needed to configure on TA gateway.

Note: this is a peer-to-peer mode to connect VoIP server and TA gateway, you also need to create a service provider trunk connecting to TA gateway on your VoIP server.

- **Port Register** – the VoIP server requires TA gateway to register with the provider using an authentication ID and password.

If you choose the VoIP server, you need to fill in User Name, Authentication Name and Password to register the FXS port.

The screenshot shows the 'Edit FXS Port - 1' configuration window. It has two tabs: 'General' and 'Other Settings'. The 'General' tab is active. Under 'General', there are two input fields: 'Caller ID Name' and 'Caller ID Number', both containing the value '301'. Below this is the 'VoIP Server Template' section, which is highlighted with a red rectangular box. Inside this box, there are four fields: 'VoIP Server' (a dropdown menu showing 'pbx(2)'), 'User Name' (text input '301'), 'Authentication Name' (text input '301'), and 'Password' (a masked text input with 8 dots). To the right of the 'Password' field is a 'From User' field, which is currently empty.

Figure 5-1 Port Register

- **Template Register** – register to your VoIP server and apply the template to FXS ports. The FXS ports will register to the server with the same account.

Figure 5-2 Template Register

Check VoIP settings below.

➤ **General**

Table 5-1 Description of VoIP Server General Settings

| Items | Description |
|------------------------------|---|
| Server ID | The ID for the VoIP server template. |
| Server Name | The name for the VoIP server template. |
| Type | Choose the type of the VoIP server, SIP or IAX. |
| Transport | This will be the transport method used by the SIP Trunk. This method is given by the SIP trunk provider. The options are UDP (default), TCP, and TLS. |
| Hostname/IP | VoIP server hostname or IP address. 5060 is the standard port number used by SIP protocol. Don't change this part if it is not required. |
| Domain | VoIP server hostname. An IP address also can be filled here. |
| Enable Outbound Proxy Server | A proxy that receives requests from a client. Even though it may not be the server resolved by the Request-URI. |
| Failover Hostname/IP | Set the failover server. This server will be used if the primary server is unavailable. |
| Register Mode | <ul style="list-style-type: none"> Port Register: ports will register to this server with individual accounts. Template Register: ports will register to this server with the |

| | |
|--|---|
| | same account. <ul style="list-style-type: none"> • Service Provider: disable registration. |
|--|---|

➤ Advanced

Table 5-2 Description of VoIP Server Advanced Settings

| Items | Description |
|------------------------------|--|
| Enable SRTP | Define if SRTP is enabled for this VoIP server. |
| Qualify | Send check alive packets to the SIP provider. |
| Caller ID | Specify the caller ID to use when making outbound calls over this VoIP server. |
| Maxmum Channels | Control the maximum number of simultaneous calls. Set as 0 to specify no maximum. |
| Realm | Realm is a string to be displayed to users so they know which username and password to use. |
| Authenticating Incoming Call | When an incoming call reaches TA device and sends INVITE packet to TA, TA responds 401, but the Realm info in 401 Response does not match the Realm set on TA VoIP Server, the provider will refuse to authenticate. If you set this option to No, TA will not reply a 401 Response to the provider to authenticate the incoming call. |
| DTMF Mode | Set default mode for sending DTMF of this trunk. Default setting: rfc2833 |
| Codec | Define the codec for this sip trunk and its priority. TA400/800 supports the following codec: u-law, a-law, G726, G729A/B, GSM, ADPCM, G722, G723, iLBC. |

Dial Pattern Template


Dial pattern template specifying how to route the calls from FXS ports to VoIP server extensions or external numbers. The number of dial pattern templates is limited by the number of ports each TA FXS Gateway model has.

The screenshot shows a dialog box titled "Edit Dial Pattern Template - DialPatternTemplate1". It contains the following fields and controls:

- Template ID: 1
- Template Name: DialPatternTemplate1
- A table with three columns: Dial Pattern, Strip, and Prepend. The Dial Pattern column contains the text "X.". There is a close button (X) to the right of the table.
- An "Add" button with a green plus sign icon.
- "Save" and "Cancel" buttons at the bottom.

Figure 5-3 Dial Pattern Template

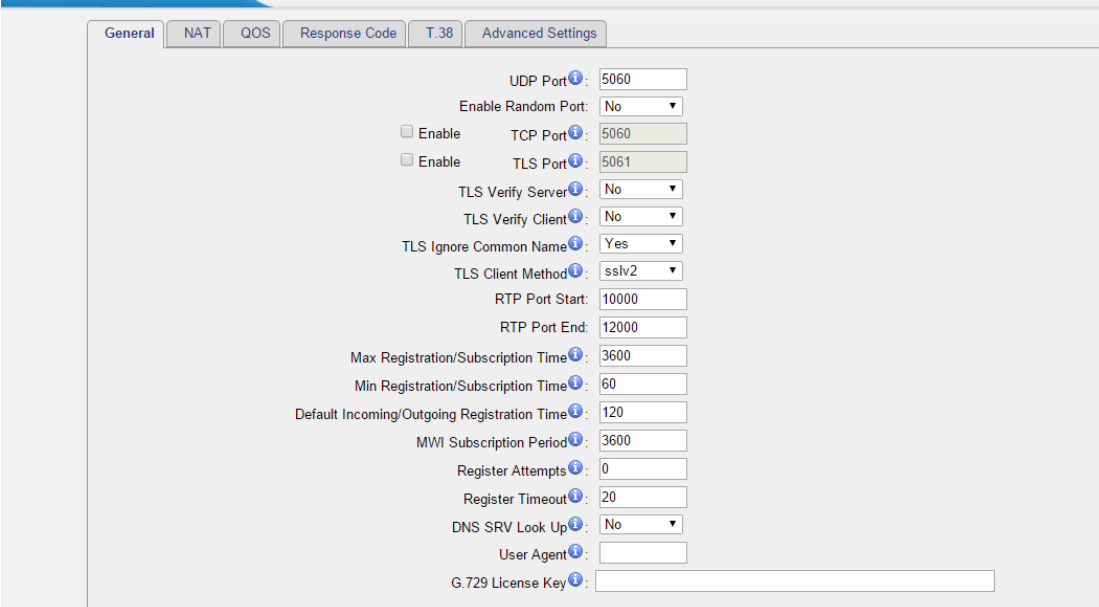
Table 5-3 Description of Dial Pattern Template Settings

| Items | Description |
|---------------|---|
| Template ID | The ID for this template. |
| Template Name | A name for this template. |
| Dial Pattern | Calls from the FXS port should match the dial pattern set on this template, or the call cannot be established. Hover the pointer over  to read tips. |
| Strip | Allows the user to specify the number of digits that will be stripped from the front of the phone number before the call is placed. |
| Prepend | The digits will be appended to the phone number before the call is placed. |
| DTMF Mode | Set default mode for sending DTMF of this trunk. Default setting: rfc2833. |
| Codec | Define the codec for this sip trunk and its priority. |

SIP Settings

It is wise to leave the default setting as provided on this page. However, for a few fields, you need to change them to suit your situation.

1) General



The screenshot displays the 'SIP Settings' configuration page, specifically the 'General' tab. The page includes several tabs: 'General', 'NAT', 'QOS', 'Response Code', 'T.38', and 'Advanced Settings'. The 'General' tab is active, showing the following settings:

- UDP Port: 5060
- Enable Random Port: No
- Enable TCP Port: 5060
- Enable TLS Port: 5061
- TLS Verify Server: No
- TLS Verify Client: No
- TLS Ignore Common Name: Yes
- TLS Client Method: sslv2
- RTP Port Start: 10000
- RTP Port End: 12000
- Max Registration/Subscription Time: 3600
- Min Registration/Subscription Time: 60
- Default Incoming/Outgoing Registration Time: 120
- MWI Subscription Period: 3600
- Register Attempts: 0
- Register Timeout: 20
- DNS SRV Look Up: No
- User Agent: (empty field)
- G.729 License Key: (empty field)

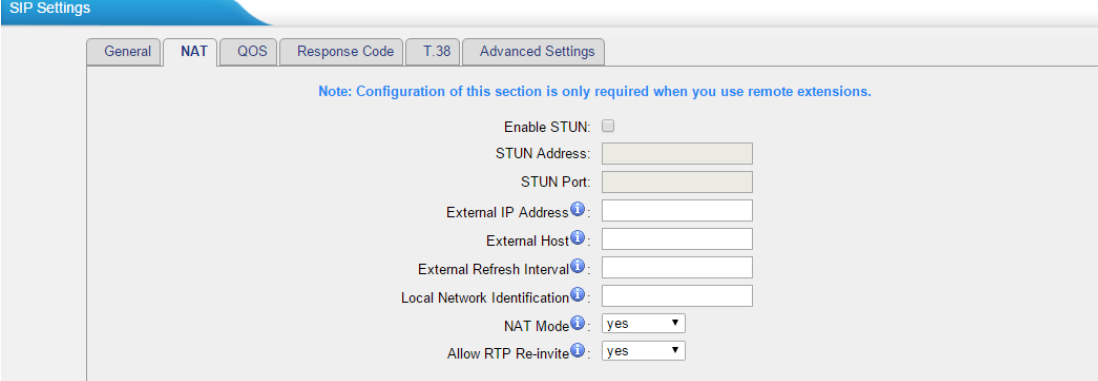
Figure 5-4 SIP General Settings

Table 5-4 Description of SIP General Settings

| Items | Description |
|---|--|
| UDP Port | Port used for SIP registrations. The default is 5060. |
| Enable Random Port | Enable or Disable Random SIP port. |
| Random Port Update Interval | Set the Random Port Update Interval. |
| TCP Port | Port used for SIP registrations. The default is 5060. |
| TLS Port | Port used for SIP registrations. The default is 5061. |
| TLS Verify Server | When using TA FXS Gateway as a TLS client, whether or not to verify server's certificate. It is "No" by default. |
| TLS Verify Client | When using TA FXS Gateway as a TLS server, whether or not to verify client's certificate. It is "No" by default. |
| TLS Ignore Common Name | Set this parameter as "No", then common name must be the same with IP or domain name. |
| TLS Client Method | When using TA FXS Gateway as TLS client, specify the protocol for outbound TLS connections. You can select it as tlsv1, sslv2 or sslv3. |
| RTP Port Start | Beginning of the RTP port range. |
| RTP Port End | End of the RTP port range. |
| DTMF Mode | Set the default mode for sending DTMF. Default setting: rfc2833 |
| Max Registration/Subscription Time | Maximum duration (in seconds) of a SIP registration. The default is 3600 seconds. |
| Min Registration/Subscription Time | Minimum duration (in seconds) of a SIP registration. The default is 60 seconds. |
| Default Incoming/Outgoing Registration Time | Default Incoming/Outgoing Registration Time: the default duration (in seconds) of incoming/outgoing registration. |
| Register Attempts | The number of SIP REGISTER messages to send to a SIP Registrar before giving up. The default is 0 (no limit). |
| Register Timeout | Number of seconds to wait for a response from a SIP Registrar before classifying the register has timed out. The default is 20 seconds. |
| Calling Channel Codec Priority | Once enabled, when dialing out via SIP/SPS trunks, the codec of calling channel will be selected preferentially. If not, TA FXS Gateway will follow the priority order in your SIP/SPS trunks. |
| Video Support | Support SIP video or no. The default is yes. |
| Max Bit Rate | Configure the max bit rate for video stream. The default: 384kb/s. |
| DNS SRV Look Up | Please enable this option when your SIP trunk contains |

| | |
|-------------------|---|
| | more than one IP address. |
| User Agent | To change the user agent parameter of asterisk, the default is "TA FXS Gateway"; you can change it if needed. |
| G.729 License Key | If you would like to use G.729 codec, please buy the license and enter the license key here. |

2) NAT



SIP Settings

General NAT QOS Response Code T.38 Advanced Settings

Note: Configuration of this section is only required when you use remote extensions.

Enable STUN:

STUN Address:

STUN Port:

External IP Address:

External Host:

External Refresh Interval:

Local Network Identification:

NAT Mode:

Allow RTP Re-invite:

Figure 5-5 NAT Settings

Table 5-5 Description of SIP General Settings

| Items | Description |
|---------------------------|--|
| Enable STUN | STUN (Simple Traversal of UDP through NATs) is a protocol for assisting devices behind a NAT firewall or router with their packet routing. |
| STUN Address | The STUN server allows clients to find out their public address, the type of NAT they are behind and the internet side port associated by the NAT with a particular local port. This information is used to set up UDP communication between the client and the VOIP provider and so establish a call. |
| External IP Address | The IP address that will be associated with outbound SIP messages if the system is in a NAT environment. |
| External Host | Alternatively you can specify an external host, and the system will perform DNS queries periodically. This setting is only required when your public IP address is not static. It is recommended that a static public IP address is used with this system. Please contact your ISP for more information. |
| External Refresh Interval | Used to identify the local network using a network number/subnet mask pair when the system is behind a NAT or firewall. Some examples of this are as follows: "192.168.0.0/255.255.0.0": All RFC 1918 addresses are local networks; "10.0.0.0/255.0.0.0": Also RFC1918; "172.16.0.0/12": Another RFC1918 with CIDR notation; "169.254.0.0/255.255.0.0": Zero conf local network. |

| | |
|--------------------|--|
| | Please refer to RFC1918 for more information. |
| NAT Mode | Global NAT configuration for the system; the options for this setting are as follows: Yes = Use NAT. Ignore address information in the SIP/SDP headers and reply to the sender's IP address/port. No = Use NAT mode only according to RFC3581. Never = Never attempt NAT mode or RFC3581 support. Route = Use NAT but do not include rport in headers. |
| Allow RTP Reinvite | By default, the system will route media streams from SIP endpoints through itself. Enabling this option causes the system to attempt to negotiate the endpoints to route packets to each other directly, bypassing the system. It is not always possible for the system to negotiate endpoint-to-endpoint media routing. |

3) Qos

QoS (Quality of Service) is a major issue in VoIP implementations. The issue is how to guarantee that packet traffic for a voice or other media connection will not be delayed or dropped due interference from other lower priority traffic. When the network capacity is insufficient, QoS could provide priority to users by setting the value.

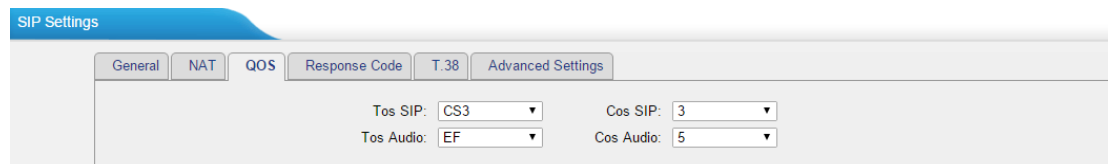


Figure 5-6 Qos

Note: It's recommended that you configure the QoS in your router or switch instead of TA FXS Gateway side.

4) Response Code

You can change the response code on TA FXS Gateway to the one you want before sending it to the VoIP server. It helps the VoIP server understand better the exact call status, like busy, no response and others.

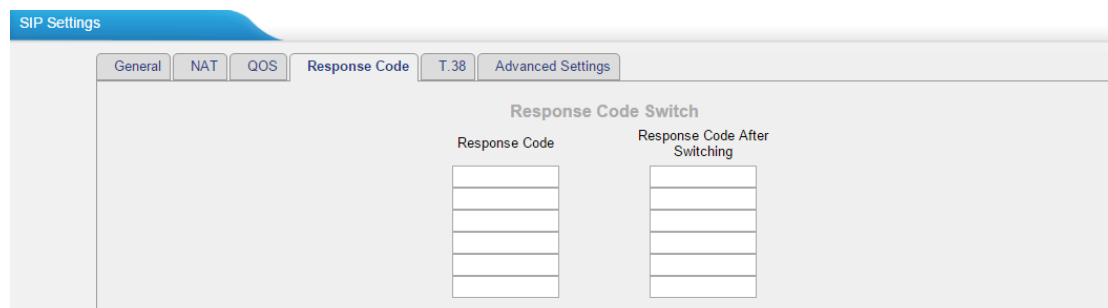


Figure 5-7 Response Code

Note: We don't recommend configuring this if you are not familiar with the code of call status from the VoIP server.

5) T.38

Settings on this page is for the purpose of improving receiving and sending T.38 FAX.

SIP Settings

General NAT QOS Response Code T.38 Advanced Settings

Re-invite SDP Not Add T.38 Attributes: No

Error Correction: Redundancy

T38 Max BitRate: 14400

T.38 Support Modems: V17 + V27ter + V29

Save Cancel

Figure 5-8 T.38 Settings

Table 5-6 Description of T.38 Settings

| Items | Description |
|---------------------------------------|---|
| Re-invite SDP Not Add T.38 Attributes | If set to Yes, SDP in re-invite packet will not add T.38 attributes. |
| Error Correction | Re-invite SDP T38FaxUdpEc. |
| T.38 Max BitRate | Set T38 Max BitRate. |
| T.38 Support Modems | Choose the T.38 modems: <ul style="list-style-type: none"> • V27ter • V29 • V17 • V27ter + V29 • V17 + V29 • V17 + V27ter + V29 |

7) Advanced Settings

SIP Settings

General NAT QOS Response Code T.38 Advanced Settings

Call ID Field: From

DID Field: To

180 Ringing:

Remote Party ID: send trust

Allow Guest: No

Pedantic: No

Alwaysauthreject: Yes

OPTIONS Response 200: Yes

Session-timers: Accept

Session-expires: 1800 s

Session-minse: 90 s

Session-refresher: Uas

Figure 5-9 SIP Advanced Settings

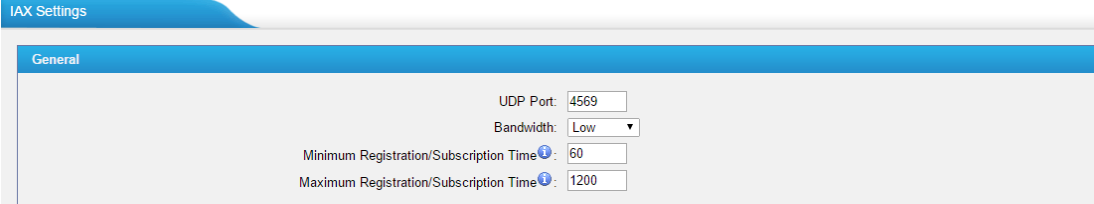
Table 5-7 Description of SIP Advanced Settings

| Items | Description |
|---------------|---|
| Call ID Field | Where to get the caller ID in SIP packet. |

| | |
|----------------------|--|
| DID Field | Where to get the DID in SIP packet. |
| 180 Ringing | It is set when the telecom provider needs. Usually it is not needed. |
| Remote Party ID | Whether to send Remote-Party-ID on SIP header or not. Default: no. |
| Allow Guest | Whether to allow anonymous registration extension or not. Default: no. It's recommended that it is disabled for security reason. |
| Pedantic | Enable pedantic parameter. Default: no. |
| Alwaysauthreject | If enabled, when TA FXS Gateway rejects "Register" or "Invite" packets, TA FXS Gateway always respond the packets using "SIP404 NOT FOUND". It's recommended that it is enabled for security reason. |
| OPTIONS Response 200 | If set to yes, the response to an OPTIONS is always 200OK. |
| Session-timers | Enable session-timer mode, default: yes. If you find the call is cut off every 15 minutes every time, please disable this. |
| Session-expires | The max refresh interval |
| Session-minse | The min refresh interval, which mustn't be shorter than 90s. |
| Session-refresher | Choose the session-refresher, the default is Uas. |

IAX Settings

IAX is the Internal Asterisk Exchange protocol, you can connect to TA FXS Gateway or register IAX trunk to another IAX server. It's supported by the asterisk-based IPPBX.



The screenshot shows the 'IAX Settings' configuration interface. Under the 'General' tab, the following settings are visible:

- UDP Port: 4569
- Bandwidth: Low
- Minimum Registration/Subscription Time: 60
- Maximum Registration/Subscription Time: 1200

Figure 5-10 IAX Settings

Table 5-8 Description of IAX Settings

| Items | Description |
|-----------|--|
| UDP Port | Port used for IAX2 registrations. Default is 4569. |
| Bandwidth | Low/medium/high with this option you can control which |

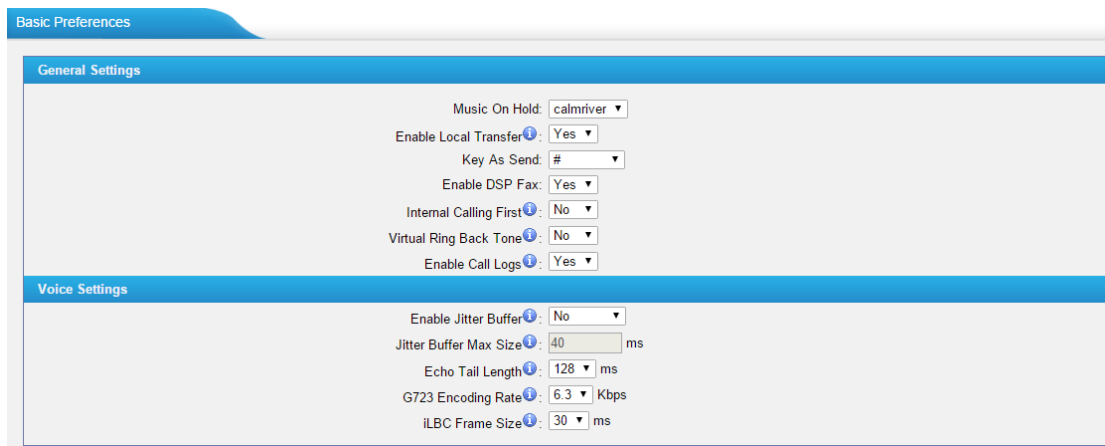
| | |
|---|--|
| | codec to be used. |
| Minimum Registration Time/Subscription Time | Minimum duration (in seconds) of an IAX2 registration. Default is 60 seconds |
| Maximum Registration Time/Subscription Time | Maximum duration (in seconds) of an IAX2 registration. Default is 1200 seconds. |

Gateway Settings

This chapter explains Gateway settings, which can be applied globally to TA400/800. The gateway settings can be configured under **Gateway**→ **Gateway Settings**.

- [Basic Preferences](#)
- [Feature Codes](#)
- [Speed Dial](#)

Basic Preferences



The screenshot shows the 'Basic Preferences' configuration page. It is divided into two main sections: 'General Settings' and 'Voice Settings'. Under 'General Settings', there are several options: 'Music On Hold' is set to 'calmriver'; 'Enable Local Transfer' is set to 'Yes'; 'Key As Send' is set to '#'; 'Enable DSP Fax' is set to 'Yes'; 'Internal Calling First' is set to 'No'; 'Virtual Ring Back Tone' is set to 'No'; and 'Enable Call Logs' is set to 'Yes'. Under 'Voice Settings', there are five options: 'Enable Jitter Buffer' is set to 'No'; 'Jitter Buffer Max Size' is set to '40 ms'; 'Echo Tail Length' is set to '128 ms'; 'G723 Encoding Rate' is set to '6.3 Kbps'; and 'iLBC Frame Size' is set to '30 ms'.

Figure 6-1 Basic Preferences

Table 6-1 Basic Preferences

| General Settings | |
|------------------------|--|
| Music On Hold | Set hold music for the system. |
| Enable Local Transfer | By default, the system will use the transfer rules set in TA400/800. If the option is set to “No”, the system will use the transfer mechanism of the SIP trunk. |
| Key As Send | Set the “#” or “*” to perform as a send key while dialing. Default is “#”. |
| Enable DSP Fax | Enable DSP to optimize Fax reception. |
| Internal Calling First | If set to “Yes”, when making calls between internal ports (through Caller ID Number), system will call the destination port directly without VoIP server, or through VoIP server as priority if set to “No”. |
| Virtual Ring Back Tone | Once enabled, when the caller dials out, the caller will only hear the virtual ring back tone generated by the system before the callee answers the call. |
| Enable Call Logs | If set to “Yes”, the system will store the call logs. If set to “No”, the call logs will not be kept anymore, but the call logs stored previously will be there. |

| Voice Settings | |
|------------------------|---|
| Enable Jitter Buffer | Forces the use of a jitter buffer on the received side of a SIP channel. The call quality will be improved if this option is enabled. |
| Jitter Buffer Max Size | Max length of the jitter buffer. Default is 40 milliseconds |
| Echo Tail Length | In some cases, the echo canceller doesn't train quickly enough and there is echo at the beginning of the call which then quickly fades out. |
| G723 Encoding Rate | Set the G723 encoding rate. |
| iLBC Frame Size | Set the iLBC packet frame size. |

Feature Codes

There are various feature codes on TA400/800. The feature codes are used to acquire the gateway info or activate and inactivate supplementary services. The default feature codes are illustrated below. The parameters for feature codes are configurable.

➤ General

Table 6-2 Description of General Feature Code

| Items | Default | Description |
|----------------------|---------|---|
| Internal Call Prefix | *99 | Dial the feature code and the FXS port number when making calls between the analog phones connected to multiple ports of the TA FXS Gateway without the use of a VoIP server. |
| Speed Dial Prefix | *98 | The prefix number for applying a speed dialing. The prefix should be added ahead of the speed dial number. |
| Attended Transfer | *3 | Users may transfer an incoming call by dialing *3 on their phone. |
| Blind Transfer | *03 | Users may blind transfer an incoming call by dialing*03 on their phone. |
| Direct IP Calling | *96 | Direct IP calling allows two parties, that is, a FXS Port with an analog phone and another VoIP Device, to talk to each other in an ad hoc fashion without a SIP proxy. The default "Direct IP Calling" feature code is *96. For detailed instruction, please refer to Basic Operation. |
| Check Number | *97 | Users can check the analog phone's number by simply dialing the "Check Number" feature code on the phone. The default "Check Number" feature code is *97. |
| Voice Menu | *** | Users may enter the voice prompt menu by pressing *** on their phone. |
| Voice Menu | 123456 | The password of voice menu is required before entering |

| | | |
|-------------------|--|--|
| Password Settings | | the advanced settings. The default password is 123456. |
|-------------------|--|--|

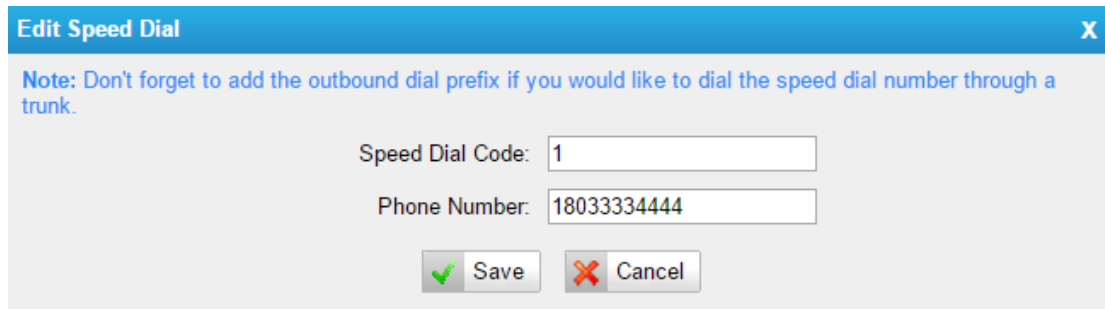
➤ **Call Forwarding Preferences**

Table 6-3 Description of Call Forwarding Preferences

| Items | Default | Description |
|---------------------------|---------|---|
| Reset to Defaults | *70 | Users may reset all call forwarding defaults by calling *70 on their phone. |
| Enable Forward All Calls | *71 | Users may enable always forward by calling *71 on their phone. |
| Disable Forward All Calls | *071 | Users may disable always forward by calling *071 on their phone. |
| Enable Forward When Busy | *72 | Users may enable busy forward by dialing *72 on their phone. |
| Disable Forward When Busy | *072 | Users may disable busy forward by calling *072 on their phone. |
| Enable Forward No Answer | *73 | Users may enable no answer forward by calling *73 on their phone. |
| Disable Forward No Answer | *073 | Users may disable no answer forward by calling *073 on their phone. |
| Forward to Internal Port | *74 | Users may activate call forwarding to port by dialing this feature code, followed by the port ID to forward all calls to. |
| Forward to Number | *75 | Users may activate call forwarding by dialing this feature code, followed by the extension or phone number to forward all calls to this number. Note: users may activate Forward to number by dialing *75 + phone number. E.g. by dialing *75501, all calls will be forwarded to extension 501. |
| Forward to Hunt Group | *76 | Users may activate call forwarding to hunt group by dialing code, followed by the hunt group ID to forward all calls to. |
| Enable Do Not Disturb | *77 | Activate "Do Not Disturb". Once activated, the FXS port will reject all incoming calls. |
| Disable Do Not Disturb | *077 | Disable "Do Not Disturb" for the FXS port by pressing the feature code on the phone. It will recover normal ringing upon the arrival of incoming calls. |

Speed Dial

Speed Dial feature is available on TA400/800 that allowing you to place a call by pressing a reduced number of keys. There are 16 configurable Speed Dial templates available on TA400/800.



Edit Speed Dial X

Note: Don't forget to add the outbound dial prefix if you would like to dial the speed dial number through a trunk.

Speed Dial Code:

Phone Number:

Figure 6-2 Speed Dial

- **Speed Dial Code**
The speed dialing number.
- **Phone Number**
The number you want to call.

To make a speed dial, e.g. you want to call 18033334444, simply dial *981. The *98 tells TA400/800 that you want to use the Speed Dial and the 1 is the Speed Dial Code for destination number 18033334444.

Note: do not forget to add the dial pattern according to the selected dial pattern template. That is the destination number should match the FXS port dial pattern.

Prompt Settings

This chapter explains prompt settings on TA400/800.

- Custom Prompts
- Music on Hold Prompts
- System Prompts

Custom Prompts

We can upload the prompts in this page; you can also download it and save it as a backup.

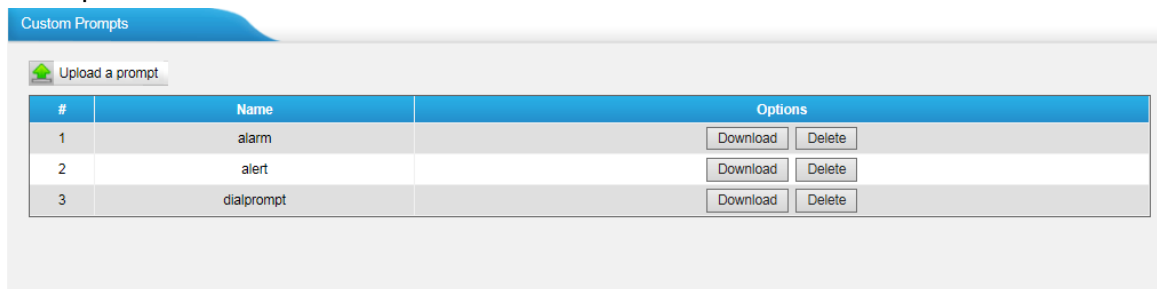


Figure 7-1 Custom Prompts

The administrator can upload prompts by doing the following:

- 1) Click "Upload Prompt".
- 2) Click "Browse" to choose the desired prompt.
- 3) Click "Upload" to upload the selected prompt.

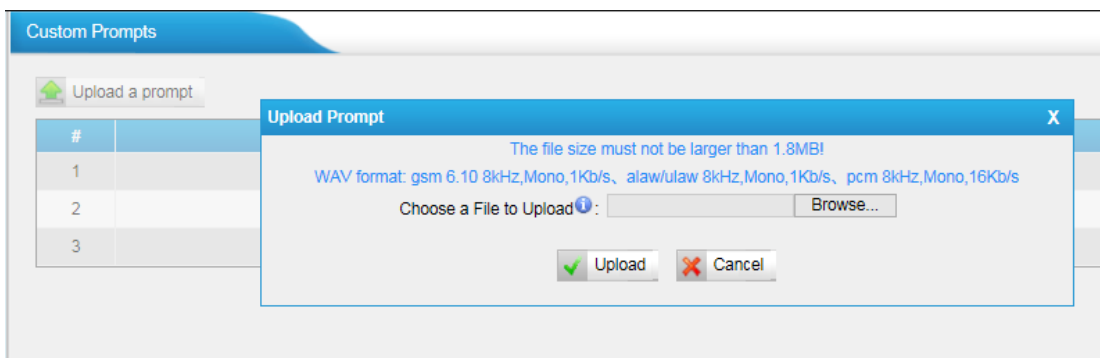


Figure 7-2 Upload A Prompt

Note: The file size must not be larger than 1.8 MB, and the file must be WAV format:
 GSM 6.10 8 kHz, Mono, 1 Kb/s;
 Alaw/Ulaw 8 kHz, Mono, 1 Kb/s;
 PCM 8 kHz, Mono, 16 Kb/s.

Music on Hold Prompts

In this page, we can upload the music on hold prompts.

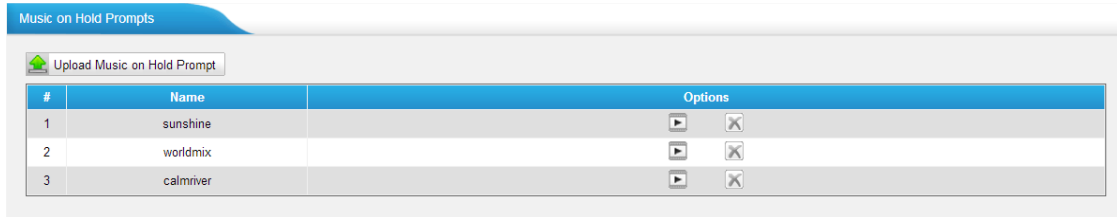


Figure 7-3 Music On Hold

The administrator can upload on hold music as follows:

- 1) Click "Upload Music on Prompt".
- 2) Click "Browse" to choose the desired audio file.
- 3) Click "Upload" to upload the selected file.

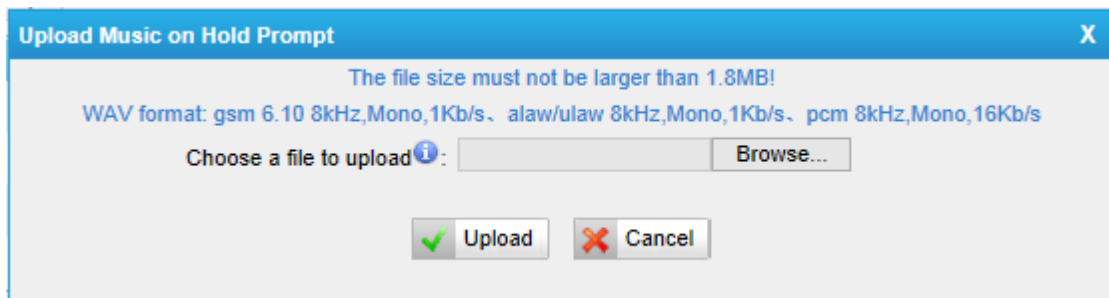


Figure 7-4 Upload Music on Hold Prompt

Note: The file size must not be larger than 1.8 MB, and the file must be WAV format:
 GSM 6.10 8 kHz, Mono, 1 Kb/s;
 Alaw/Ulaw 8 kHz, Mono, 1 Kb/s;
 PCM 8 kHz, Mono, 16 Kb/s.

System Prompts

TA FXS Gateway supports Chinese and English system prompts. You can download the appropriate language you need.

Notes:

1. Auto-detection is highly recommended. But if you prefer to download via HTTP or TFTP server, please contact the local dealer for the prompts.
2. When update successfully, just click "Apply Changes" on Web then it will take effect, there is no need to reboot.

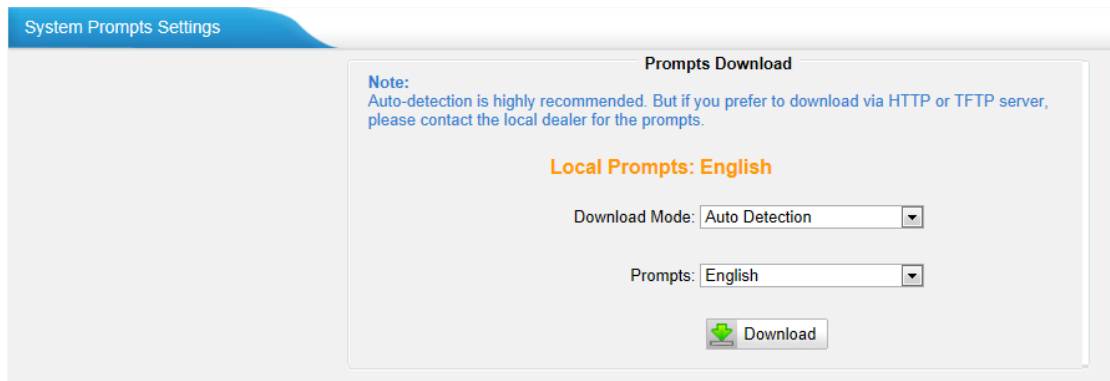


Figure 7-5 System Prompts Settings Page

Advanced Settings

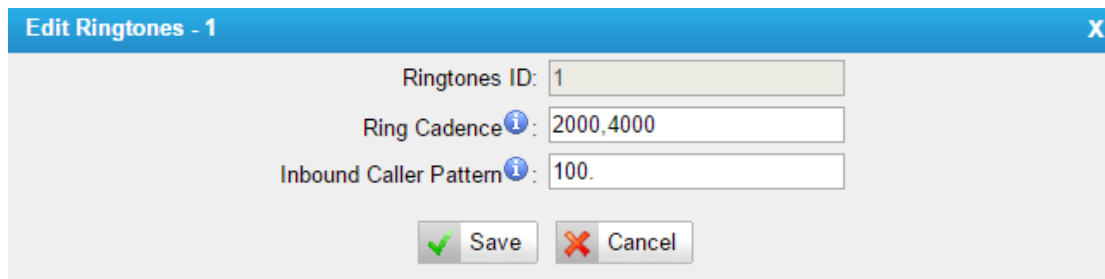
This chapter explains SIP settings and Distinctive Ringtones.

- [Distinctive Ringtones](#)
- [Tone Zone Settings](#)
- [RADIUS Settings](#)

Distinctive Ringtones

TA400/800 provides 10 configurable distinctive ringtones. Users could configure different ringtones to match different incoming caller ID.

For example, if Inbound Caller Pattern is configured as “100.”, all the incoming calls start with digits “100” will ring using cadence “2000,4000” as the following figure shows.

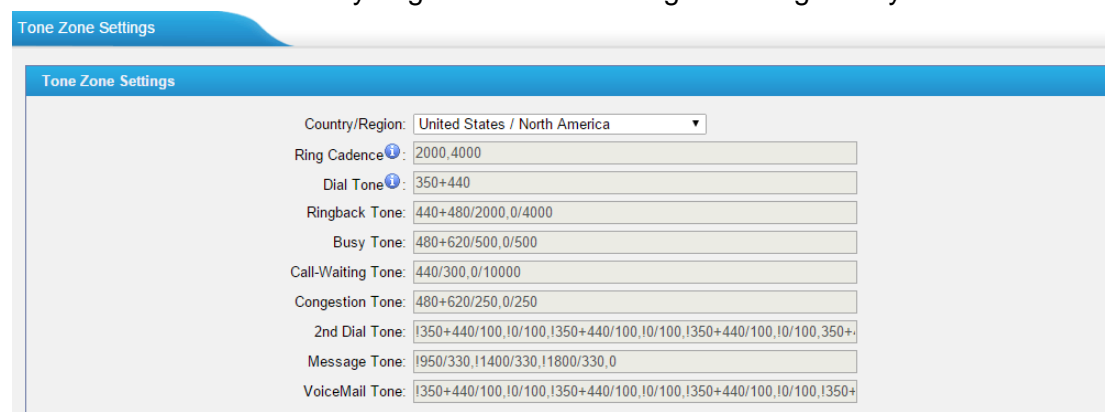


The screenshot shows a dialog box titled "Edit Ringtones - 1". It contains three input fields: "Ringtones ID" with the value "1", "Ring Cadence" with the value "2000,4000", and "Inbound Caller Pattern" with the value "100.". Below the fields are two buttons: "Save" (with a green checkmark icon) and "Cancel" (with a red X icon).

Figure 8-1 Distinctive Ringtones

Tone Zone Settings

Advanced ring tones for all the FXS ports can be configured on this page. There are pre-programmed tone zone settings for some countries and regions. Users can simply find and select their country to get tone zone settings for the gateway.



The screenshot shows the "Tone Zone Settings" page. At the top, there is a dropdown menu for "Country/Region" set to "United States / North America". Below this are several input fields for different tones:

- Ring Cadence: 2000,4000
- Dial Tone: 350+440
- Ringback Tone: 440+480/2000,0/4000
- Busy Tone: 480+620/500,0/500
- Call-Waiting Tone: 440/300,0/10000
- Congestion Tone: 480+620/250,0/250
- 2nd Dial Tone: 1350+440/100,10/100,1350+440/100,10/100,1350+440/100,10/100,350+
- Message Tone: 1950/330,11400/330,11800/330,0
- VoiceMail Tone: 1350+440/100,10/100,1350+440/100,10/100,1350+440/100,10/100,1350+

Figure 8-2 Tone Zone Settings

Users may also configure the tone zone according to the national standard by selecting "User custom for Tone Zone". Please refer to the document below and

configure the tone zone settings on TA FXS Gateway:

<http://www.itu.int/ITU-T/inr/forms/files/tones-0203.pdf>

Figure 8-3 Customize Tones

Table 8-1 Description of Tone Zone Settings

| Items | Description |
|-------------------|---|
| Country/Region | Choose the country to get pre-programmed tone zone settings or choose "User custom for Tone Zone" to configure the settings manually. |
| Ring Cadence | Configuration option for all FXS ports ring cadence for all incoming calls. |
| Dial Tone | Prompt tone of off-hook dial tone. |
| Ringback Tone | The tone sent to caller when ringing is on. |
| Busy Tone | Used for busy line prompt. |
| Call-Waiting Tone | Used for notification in call waiting. |
| Congestion Tone | Used to indicate that an invalid code has been dialed, or that all circuits (trunks) are busy and/or the call is unroutable. |
| 2nd Dial Tone | Used for the second stage dial tone. |
| Message Tone | Used for message tone. |
| VoiceMail Tone | Used for voicemail tone. |

RADIUS Settings

TA FXS Gateways supports RADIUS (Remote Authentication Dial In User Service) protocol. RADIUS feature is mainly for billing purpose on TA FXS Gateway. There are primary and secondary RADIUS server configurations available. Once the primary server is unreachable, the RADIUS requests will be sent to the secondary server.

RADIUS Preferences

General Settings

Enable RADIUS

Primary Server IP

Primary Server Port

Primary Server Key

Secondary Server IP

Secondary Server Port

Secondary Server Key

RADIUS Timeout s

RADIUS Retry Times

Call Out CDR

Call In CDR

No Answer CDR

Figure 8-4 RADIUS Settings

Table 8-2 Description of RADIUS Settings

| Items | Description |
|-----------------------|--|
| Enable RADIUS | Enable RADIUS on TA FXS Gateway. |
| Primary Server IP | Set IP address of the primary server. |
| Primary Server Port | Default is 1813. Specifies the port to be used for the primary RADIUS account. |
| Primary Server Key | Specifies the key to be used to authenticate the RADIUS connection to the Primary server. The key is set according to the RADIUS server. |
| Secondary Server IP | Set IP address of the primary server. The second sever will be activated the primary one becomes unusable. |
| Secondary Server Port | Default is 1813. Specifies the port to be used for the second RADIUS account. |
| Secondary Server Key | Specifies the key to be used to authenticate the RADIUS connection to the second server. The key is set according to the RADIUS server. |
| RADIUS Timeout | Specifies the number of seconds to wait for a response after the RADIUS message is sent to the server. Default: 10 seconds. The retransmission will be performed if there is no response after the timeout. |
| RADIUS Retry Times | Specifies the number of times the RADIUS messages will send to the RADIUS server before giving up. Default: 3. |
| Call Out CDR | Whether to send "Call Out CDR" to RADIUS server or not. |
| Call In CDR | Whether to send "Call In CDR" to RADIUS server or not. |
| No Answer CDR | Whether to send "Call Out CDR" to RADIUS server or not. |

Network Preferences



This chapter explains network settings on TA400/800. Click the main menu **System** on the top of the Web GUI to check the network settings.

- [LAN Settings](#)
- [Service](#)
- [VLAN Settings](#)
- [VPN Settings](#)
- [DDNS Settings](#)
- [Static Route](#)
- [TR-069 Settings](#)

LAN Settings

After successfully logging in the TA400/800 Web GUI for the first time, users could go **System**→**Network Preferences**→**LAN Settings** to configure the network for TA400/800.

The screenshot shows the 'LAN Settings' page with the following configuration:

- Hostname: TA400
- Mode: Static IP Address
- IP Address: 192.168.6.218
- Subnet Mask: 255.255.255.0
- Gateway: 192.168.6.1
- Primary DNS: 192.168.1.1
- Secondary DNS: (empty)
- IP Address2: (empty)
- Subnet Mask2: (empty)

Figure 9-1 LAN Settings

Table 9-1 LAN Settings

| Items | Description |
|-------------|---|
| Hostname | Set the host name for TA400/800. |
| Mode | Choose the network mode: <ul style="list-style-type: none"> • Static IP Address • DHCP • PPPoE |
| IP Address | Set the IP Address for TA400/800. |
| Subnet Mask | Set the subnet mask for TA400/800. |
| Gateway | Set the gateway for TA400/800. |
| Primary DNS | Set the primary DNS for TA400/800. |

| | |
|---------------|---|
| Secondary DNS | Set the secondary DNS for TA400/800. |
| IP Address2 | Set the second IP Address for TA400/800. |
| Subnet Mask2 | Set the second subnet mask for TA400/800. |

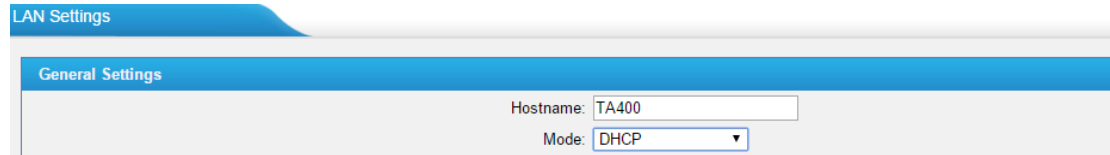


Figure 9-2 DHCP Mode

Select DHCP mode to get network automatically from the local network.

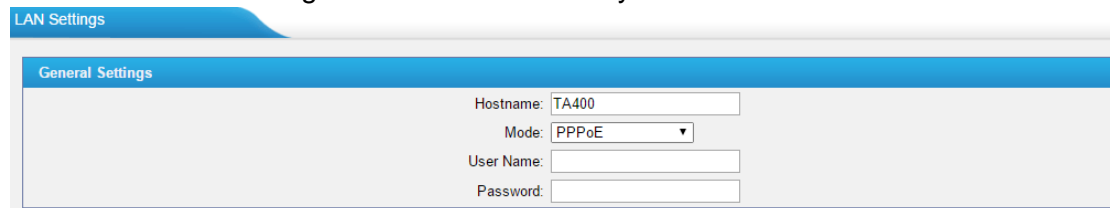


Figure 9-3 PPPoE

Fill in user name and password to access the Internet via PPPoE.

Service

The administrator can manage all the access methods on TA on the "Service" page.

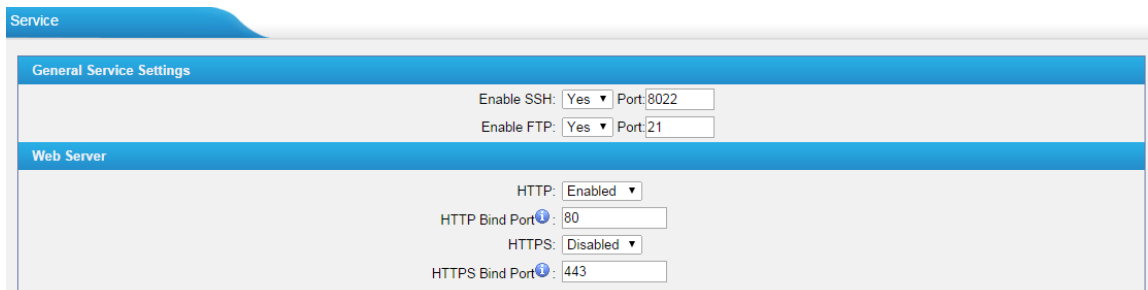


Figure 9-4 Service Settings

Table 9-2 Description of Service Settings

| Items | Description |
|-------|--|
| SSH | By using SSH, you can log in to TA400/800 and run commands. It's disabled by default. We don't recommend enabling it if not needed. The default port for SSH is 8022. |
| FTP | FTP access; The default port is 21. |
| HTTP | HTTP web access; The default port is 80. |

HTTPS

HTTPS web access, it is disabled by default, and you can enable it to get safer web access.

VLAN Settings

VLAN (Virtual Local Area Network) is a group of hosts with a common set of requirements, which communicate as if they were attached to the same broadcast domain, regardless of their physical location.

A VLAN is a broadcast domain created by switches. This means the VLAN is configured on switches, layer 3 switches. Note that some of the switches don't support VLAN.

Note:

TA400/800 acts as a VLAN client, a 3-layer switch is needed.

The screenshot shows a web-based configuration interface for VLAN settings. The title bar reads 'VLAN Settings'. Below it, the main heading is 'VLAN Over LAN'. The interface is divided into two sections, 'NO.1' and 'NO.2', each with a checkbox to enable the VLAN. Under each section, there are four input fields: 'VLAN Number', 'VLAN IP Address', 'VLAN Subnet Mask', and 'Default Gateway'. At the bottom of the form, there are two buttons: 'Save' (with a green checkmark icon) and 'Cancel' (with a red X icon).

Figure 9-5 VLAN Settings

Please follow the steps below to set up VLAN on TA400/800.

Step1. Create VLANs on your switch.

Step2. Allocate a VLAN ID and IP address for TA400/800.

Step3. Configure VLAN settings page on TA400/800.

VPN Settings

A virtual private network (VPN) is a method of computer networking typically using the public internet that allows users to privately share information between remote locations, or between a remote location and a business' home network. A VPN can provide secure information transport by authenticating users, and encrypting data to prevent unauthorized persons from reading the information transmitted. The VPN can be used to send any kind of network traffic securely. TA400/800 supports OpenVPN.

VPN Settings

General Settings

Enable VPN:

Import VPN Config: Browse...

Import

Save Cancel

Figure 9-6 VPN Settings

- **Enable VPN**
Enable VPN feature.
- **Import VPN Config**
Import configuration file of OpenVPN.

Notes:

1. Uncomment “user” and “group” in the “config” file. You can get the config package from the OpenVPN provider.
2. TA400/800 works as VPN client mode only.

DDNS Settings

DDNS(Dynamic DNS) is a method/protocol/network service that provides the capability for a networked device, such as a router or computer system using the Internet Protocol Suite, to notify a Domain Name System (DNS) name server to change, in real time, the active DNS configuration of its configured hostnames, addresses or other information.

DDNS Settings

General Settings

Note: DDNS allows you to access your network using domain names instead of IP address. The service manages changing IP address and updates your domain information dynamically. You must sign up for service through dyndns.org, freedns.afraid.org, www.no-ip.com, www.zoneedit.com

DDNS is not running

Enable DDNS:

DDNS Server:

User Name:

Password:

Host Name:

Save Cancel

Figure 9-7 DDNS Settings

Table 9-3 Description of DDNS Settings

| Items | Description |
|-------------|---|
| DDNS Server | Select the DDNS server you sign up for service. |
| User Name | User name the DDNS server provides you. |
| Password | User account's password. |

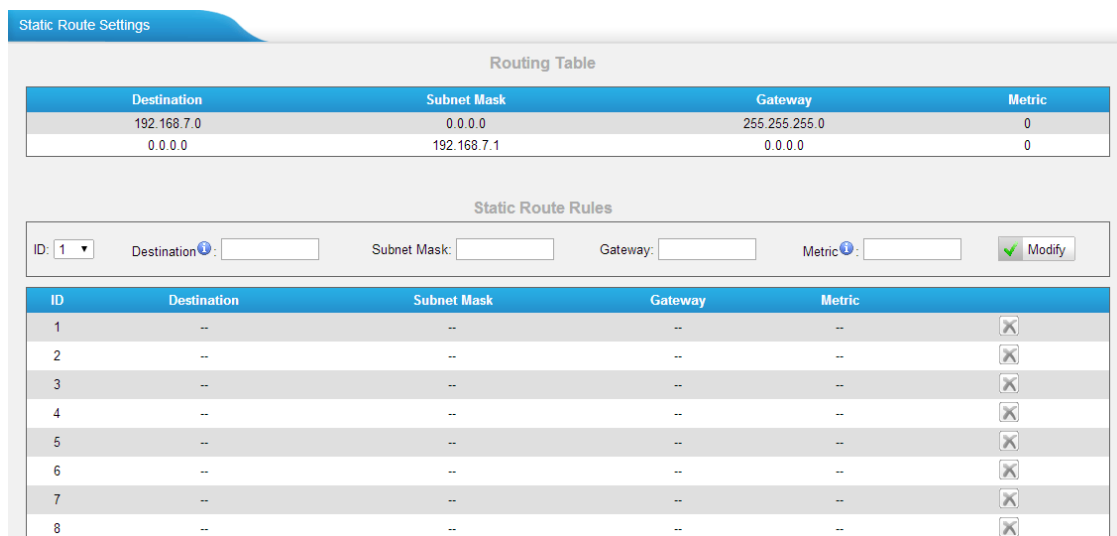
| | |
|-----------|---|
| Host Name | The host name you have got from the DDNS server |
|-----------|---|

Note: DDNS allows you to access your network using domain names instead of IP address. The service manages changing IP address and updates your domain information dynamically. You must sign up for service through dyndns.org, freedns.afraid.org, www.no-ip.com, www.zoneedit.com.

Static Route

TA FXS Gateway will have more than one Internet connection in some situations but it has only one default gateway. You will need to set some Static Route for TA FXS Gateway to force it to go out through different gateway when accessing to different internet.

The default gateway priority of TA FXS Gateway from high to low is VPN/VLAN → LAN port.



Static Route Settings

Routing Table

| Destination | Subnet Mask | Gateway | Metric |
|-------------|-------------|---------------|--------|
| 192.168.7.0 | 0.0.0.0 | 255.255.255.0 | 0 |
| 0.0.0.0 | 192.168.7.1 | 0.0.0.0 | 0 |

Static Route Rules

ID: 1 Destination: Subnet Mask: Gateway: Metric: Modify

| ID | Destination | Subnet Mask | Gateway | Metric |
|----|-------------|-------------|---------|--------|
| 1 | -- | -- | -- | -- |
| 2 | -- | -- | -- | -- |
| 3 | -- | -- | -- | -- |
| 4 | -- | -- | -- | -- |
| 5 | -- | -- | -- | -- |
| 6 | -- | -- | -- | -- |
| 7 | -- | -- | -- | -- |
| 8 | -- | -- | -- | -- |

Figure 9-8 Static Route

1) Route Table

The current route rules of TA FXS Gateway.

2) Static Route Rules

You can add new static route rules here.

Table 9-4 Description of Static Route Settings

| Items | Description |
|-------------|---|
| Destination | The destination network to be accessed to by TA FXS Gateway. |
| Subnet Mask | Specify the destination network portion. |
| Gateway | Define which gateway TA FXS Gateway will go through when accessing the destination network. |
| Metric | The cost of a route is calculated by using what are called routing |

| | |
|-----------|--|
| | metric. Routing metrics are assigned to routes by routing protocols to provide measurable statistic which can be used to judge how useful (how low cost) a route is. |
| Interface | Define which internet port to go through. |

TR-069 Settings

TR-069 (Technical Report 069) is a Broadband Forum (formerly known as DSL Forum) technical specification entitled CPE WAN Management Protocol (CWMP). It defines an application layer protocol for remote management of end-user devices. As a bidirectional SOAP/HTTP-based protocol, it provides the communication between customer-premises equipment (CPE) and Auto Configuration Servers (ACS). Yeastar TA400/800 acts as a CPE, enable TR-069 and configure the parameters in the corresponding fields, if configured correctly, the gateway will establish the TR-069 session with the ACS.

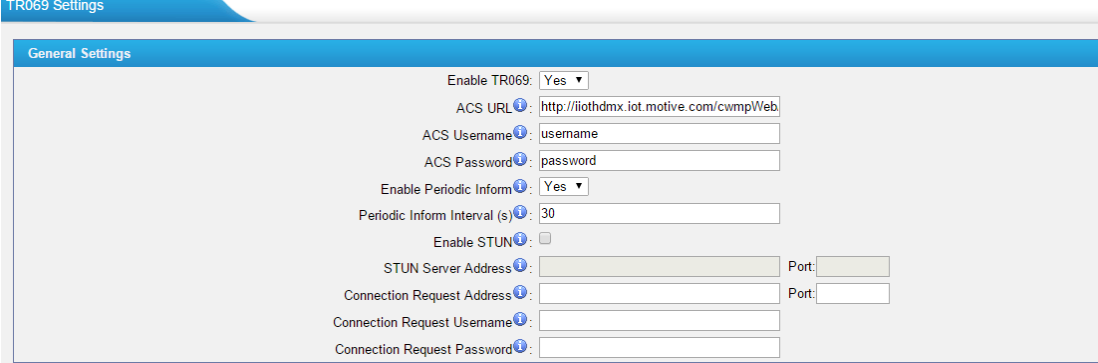


Figure 9-9 TR-069 Settings

Table 9-5 Description of TR-069 Settings

| Items | Description |
|-----------------------------|---|
| ACS URL | URL of the TR-069 ACS. This is the URL the Yeastar CPEs will send TR-069 messages to. |
| ACS Username | Username for HTTP authentication against the ACS. Leave it blank if no ASC username. |
| ACS Password | Password for HTTP authentication against the ACS. Leave it blank if no ASC password. |
| Enable Periodic Inform | Set the Yeastar device to start a new session after "Periodic Inform Interval". |
| Periodic Inform Interval(s) | Time in seconds where the CPE will initiate a new session against the ACS. |
| Enable STUN | Enable this option to discover the CPE's public IP address if it is located behind a NAT. |
| STUN Server Address/Port | The STUN Server address and port. |

| | |
|-----------------------------|--|
| STUN Server Username | The STUN Server username. |
| STUN Server Password | The STUN server password. |
| Connection Request Address | HTTP URL for an ACS to make a Connection Request notification to the CPE. |
| Connection Request Username | Username used to authenticate an ACS making a Connection Request to the CPE. Leave it blank if no username. |
| Connection Request Password | Password used to authenticate an ACS making a Connection Request to the CPE. Leave it blank if no password. |

Security Center

This chapter describes how to secure your TA400/800. It is strongly recommended that users configure firewall and other security options on TA400/800 to prevent the attack fraud and the system failure or calls loss.

- [Security Center](#)
- [Alert Settings](#)
- [AMI Settings](#)
- [Certificates](#)
- [Firewall Rules](#)
- [IP Blacklist](#)

Security Center

All the security settings including Firewall, Service, Port Settings in TA400/800 are displayed in Security Center. Users could rapidly check and configure the relevant security settings here.

1) Firewall

In the “Firewall” tab, users could check firewall configuration and alert settings. By clicking the relevant button, you can enter the configuration page directly.

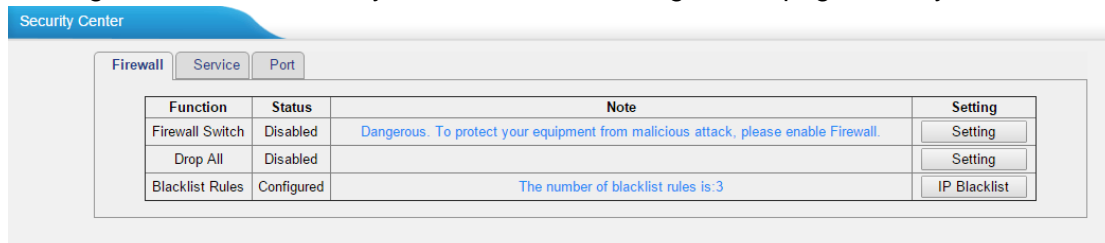


Figure 10-1 Security Center—Firewall

2) Service

In “Service” tab, you can check AMI/SSH status. For AMI/SSH, you can enter the according page by clicking the button in “Setting” column.

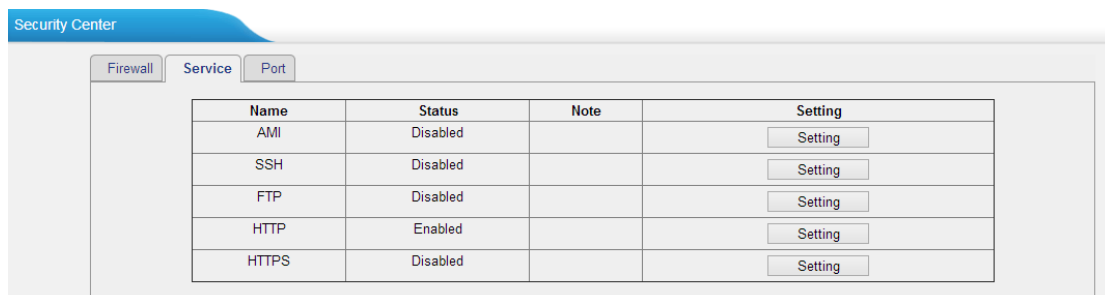
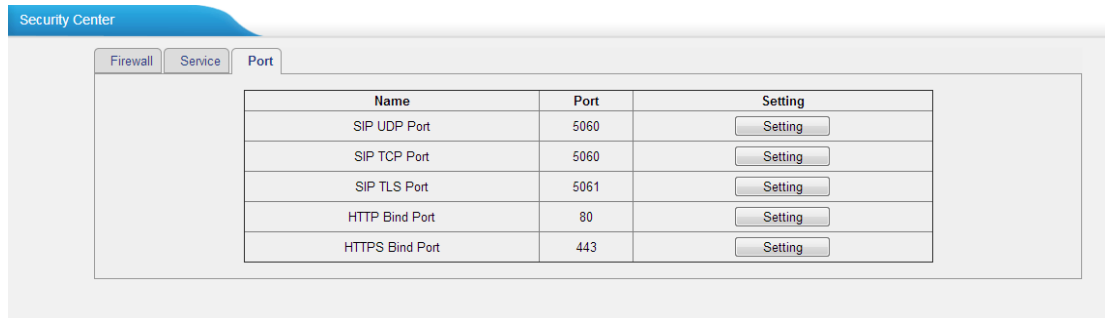


Figure 10-2 Security Center—Service

3) Port

In “Port” tab, you can check SIP port and HTTP port. You can also enter the relevant page by clicking the button in “Setting” column.



| Name | Port | Setting |
|-----------------|------|--|
| SIP UDP Port | 5060 | <input type="button" value="Setting"/> |
| SIP TCP Port | 5060 | <input type="button" value="Setting"/> |
| SIP TLS Port | 5061 | <input type="button" value="Setting"/> |
| HTTP Bind Port | 80 | <input type="button" value="Setting"/> |
| HTTPS Bind Port | 443 | <input type="button" value="Setting"/> |

Figure 10-3 Security Center—Port

Alert Settings

If the device is under attack, the system will alert users via call or E-mail. The attack modes include IP attack and Web Login.

- IPATTACK**
 When the system is attacked by IP address, the firewall will add the IP to auto IP Blacklist and notify the user if it matches the protection rule.
- WEBLOGIN**
 Web Login Alert Notification: entering the wrong password consecutively for five times when logging in TA FXS Gateway Web interface will be deemed as an attack, the system will limit the IP login within 10 minutes and notify the user.

Figure 10-4 Alert Settings

Table 10-1 Description of Alert Settings

| Phone Notification Settings | |
|-----------------------------|--|
| PHONE Notification | Whether to enable phone notification or not. |
| Number | The numbers could be set for alert notification; users can setup multiple extension and outbound phone numbers. Please separate them by “;”. Example: “500;9911”, if the extension has configured Follow Me Settings, the call would go to the forwarded number directly. |
| Attempts | The attempts to dial a phone number when there is no answer. |
| Interval | The interval between each attempt to dial the phone number. Must be longer than 3 seconds, the default value is 60 seconds. |
| Prompt | Users will hear the prompt while receiving the phone notification. |
| Email Notification Settings | |
| E-mail Notification | Whether to enable E-mail Notification or not. |

| | |
|------------------|---|
| Recipient's Name | The recipients for the alert notification, and multiple email addresses are allowed, please separate them by “;”. E.g. jerry@yeastar.com;jason@yeastar.com,456@sina.com |
| Subject | The subject of the alert email. |
| Email Content | Text content supports predefined variables. Variable names and corresponding instructions are as follows: gateway hostname:\$(HOSTNAME) attack source ip address:\$(SOURCEIP) attack dest mac:\$(DESTMACH) attack source port:\$(DESTPORT) attack source protocol:\$(PROTOCOL) attack occurred:\$(DATETIME) |

AMI Settings

The Asterisk Manager Interface (AMI) is a system monitoring and management interface provided by Asterisk. It allows live monitoring of events that occur in the system, as well enabling you to request that Asterisk perform some action. The actions that are available are wide-ranging and include things such as returning status information and originating new calls. Many interesting applications have been developed on top of Asterisk that take advantage of the AMI as their primary interface to Asterisk.

There are two main types of messages on the Asterisk Manager Interface: manager events and manager actions.

The 3rd party software can work with TA400/800 using AMI interface. It is disabled by default. If necessary, you can enable it.

Figure 10-5 AMI Settings

- User Name, Password & Port**
 After enabling AMI, you can use this username and password to log in TA400/800. The default port is 5038.

- **Permitted "IP address/Subnet mask"**
You can set which IP is allowed to log in TA400/800 AMI interface.

Certificates

TA400/800 supports TLS transport, you can configure FXS port with TLS transport. To use TLS, you should upload certificates first.

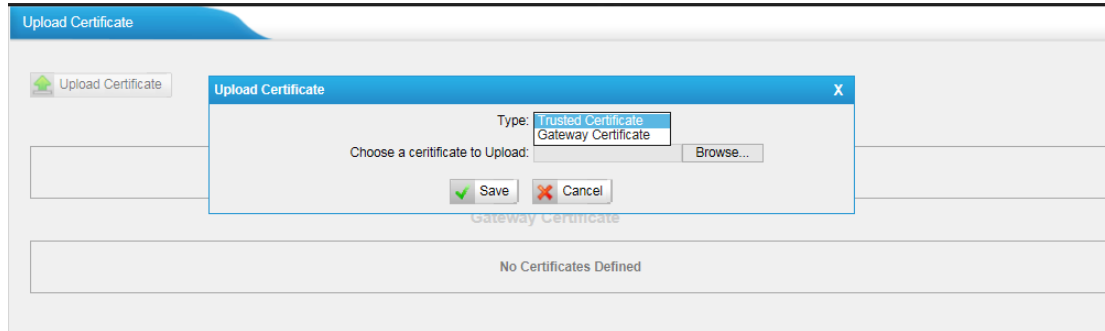


Figure 10-6 Upload Certificate

- **Trusted Certificate**
This certificate is a CA certificate. When selecting "TLS Verify Client" as "Yes", you should upload a CA. The relevant VoIP provider should also have this certificate.
- **Gateway Certificate**
This certificate is server certificate. No matter selecting "TLS Verify Client" as "Yes" or "NO", you should upload this certificate to TA400/800. If the VoIP provider enables "TLS Verify server", you should also upload the relevant CA certificate on the VoIP provider.

Firewall Rules

Firewalls are used to prevent unauthorized Internet users from accessing private networks connected to the Internet, especially intranets. All messages entering or leaving the intranet pass through the firewall, which examines each message and blocks those that do not meet the specified security criteria.



Figure 10-7 Firewall Settings

1) General Settings

Table 10-2 Description of Firewall General Settings

| Items | Description |
|-----------------|--|
| Enable Firewall | Enable the firewall to protect the device. |
| Disable Ping | Enable this item to drop net ping from remote hosts. |
| Drop All | When you enable “Drop All” feature, the system will drop all packets or connection from other hosts if there are no other rules defined. To avoid locking the devices, at least one “TCP” accept common rule must be created for port used for SSH access, port used for HTTP access and port sued for CGI access. |

2) Common Rules

There is no default rule; you can create one as required.

Figure 10-8 Common Rules

Table 10-3 Description of Common Rules

| Items | Description |
|-------------|---|
| Name | A name for this rule, e.g. "HTTP". |
| Description | Simple description for this rule. E.g. accept the specific host to access the Web interface for configuration. |
| Protocol | The protocols for this rule. |
| Port | Initial port should be on the left and end port should be on the right. The end port must be equal to or greater than start port. |
| IP | The IP address for this rule. The format of IP address is: IP/mask E.g. 192.168.5.100/255.255.255.255 for IP 192.168.5.100 E.g. 192.168.5.0/255.255.255.0 for IP from 192.168.5.0 to 192.168.5.255. |
| MAC Address | The format of MAC Address is XX:XX:XX:XX:XX:XX, X means 0~9 or A~F in hex, the A~F are not case sensitive. |
| Action | Accept: Accept the access from remote hosts. Drop: Drop the access from remote hosts. Ignore: Ignore the access. |

Note: the MAC address will be changed when it's a remote device, so it will not be working to filter using MAC for remote devices.

3) Auto Defense

Figure 10-9 Auto Defense

Table 10-4 Description of Auto Defense

| Items | Description |
|----------|--|
| Port | The port you want to auto defense, for example, 8022. |
| Protocol | Select the protocol. You can select UDP or TCP. |
| Rate | The maximum packets or connections can be handled per unit time. For example, if you configure it as below: Port: 8022 Protocol: TCP Rate: 10/min Then, it means maximum 10 TCP connections can be handled in 1 minute. The 11 th connection will be dropped. |

IP Blacklist

You can set some packets accept speed rules here. When an IP address, which hasn't been accepted in common rules, sends packets faster than the allowed speed, it will be set as a black IP address and be blocked automatically.

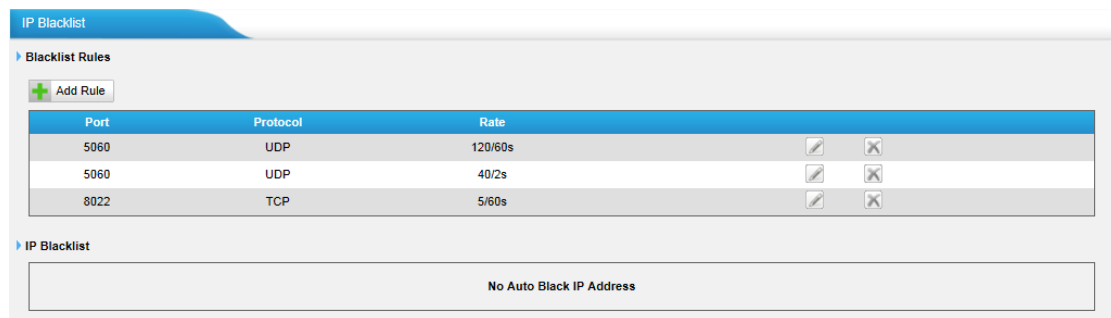


Figure 10-10 IP Blacklist Settings Page

1) Blacklist rules

We can add the rules for IP blacklist rate as demanded.

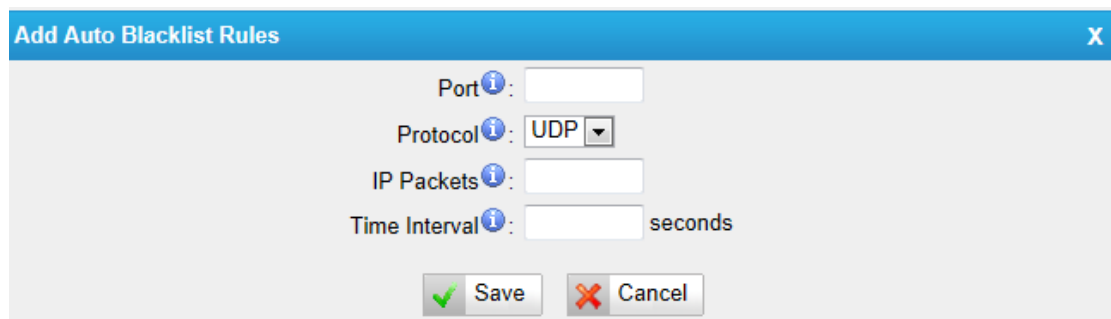


Figure 10-11 Add Blacklist Rule

Table 10-5 Description of Auto Blacklist Rules

| Items | Description |
|---------------|--|
| Port | Auto defense port |
| Protocol | Auto defense protocol. TCP or UDP. |
| IP Packets | Allowed IP packets number in the specific time interval. |
| Time interval | The time interval to receive IP packets. For example, IP packets 90, time interval 60 means 90 IP packets are allowed in 60 seconds. |

2) IP blacklist

The blocked IP address will display here, you can edit or delete it as you wish.

System Preferences

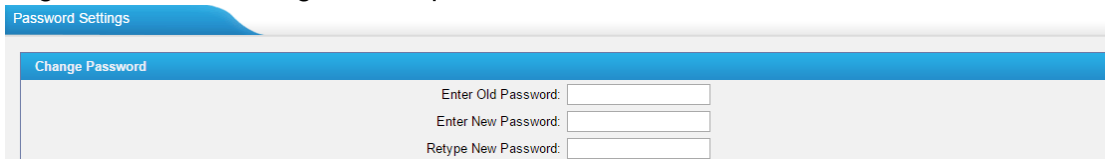
This chapter describes system maintenance settings including the followings:

- Password Settings
- Date and Time
- Auto Provision Settings
- Firmware Update
- Backup and Restore
- Reset and Reboot

Password Settings

It is highly recommended to change the system's password after first login. Go to **System**→**System Preferences**→**Password Settings** to change the password.

1. Enter the old password first.
2. Enter a new password and retype the new password to confirm. The password complexity will be detected, which will help users to set a strong password and make TA400/800 safer. A strong password is comprised of letters, numbers and characters.
3. Save the changes, the user will be automatically logged out.
4. Log in TA400/800 using the new password.



The screenshot shows a web interface for changing a password. At the top, there is a blue header with the text "Password Settings". Below this is a white box with a blue border and a blue header that says "Change Password". Inside this box, there are three input fields: "Enter Old Password:", "Enter New Password:", and "Retype New Password:". Each field has a small eye icon to its right, likely for toggling password visibility.

Figure 11-1 Password Settings

Date and Time

Please adjust the time of TA400/800 (including the time zone) consistent with your local time. Go to **System**→**System Preferences**→**Date and Time** to configure the system date and time.

Date & Time

General Settings

Server Time: Tue May 05 22:28:17 2015

Time Zone: -8 United States - Pacific Time

Daylight Saving Time: Disabled

Automatically Synchronize With an Internet Time Server
NTP Server: pool.ntp.org

Set Date & Time Manually

Date:

Time: : : AM

Figure 11-2 Date and Time

- Time Zone**
 Select your current and correct time zone on TA400/800.
- Daylight Saving Time**
 The option is disabled by default. Enable it when necessary.
- Automatically Synchronize with an Internet Time Server**
 TA400/800 will adjust its internal clock to a central network server. Please note the TA400/800 should be able to access to the Internet if you choose this method.
- Set Date & Time Manually**
 Enter the time using the numbers on your keyboard.

Note: you have to reboot the system to make the changes take effect.

Auto Provision Settings

Three methods are supported for Auto Provision: PNP, DHCP and you can manually configure a server URL to get the configuration file from the server. Go to **System**→**System Preferences**→**Auto Provision Settings** to configure.

Provision Method: _____

PNP: Yes

DHCP: No

Server URL: No

Figure 11-3 Auto Provision Methods

PNP and **DHCP** modes work along with MyPBX "TA Provisioning". Firstly, users need to configure TA400/800 on MyPBX "TA Provisioning" page. Then TA400/800 will find and get the configuration file from MyPBX during boots up.

In **PNP** mode, you just need to place the TA400/800 in the same IP range network with MyPBX, then you can find the TA400/800 and provision it on MyPBX "TA Provisioning" page.

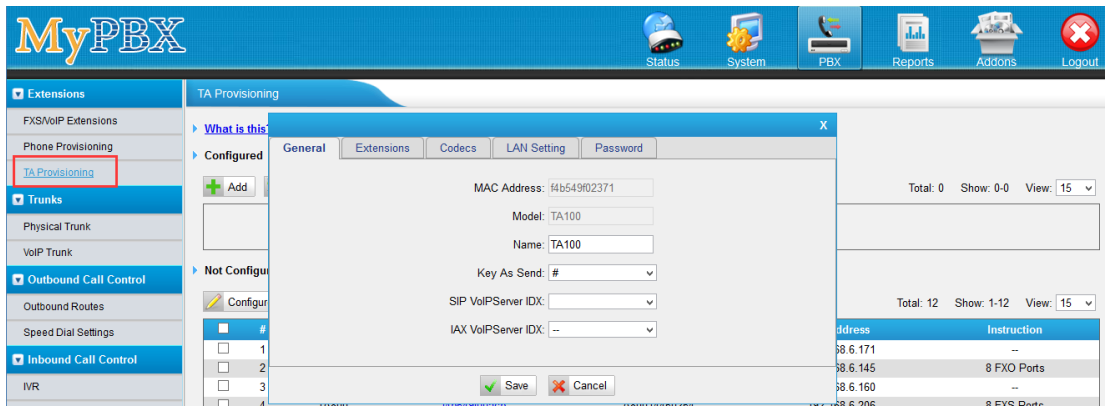


Figure 11-4 MyPBX TA Provisioning

If you use **DHCP** mode to do auto provision, you should enable DHCP Server on MyPBX to make it as a DHCP server. (System→Network Preferences→DHCP Server).

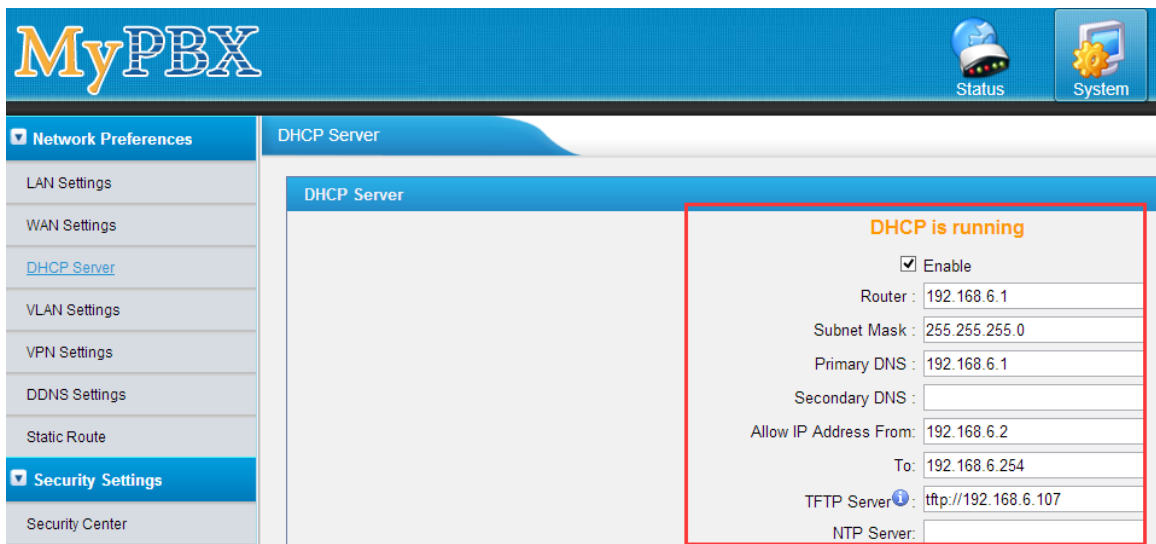


Figure 11-5 Set MyPBX as a DHCP Server

Then select DHCP mode on LAN settings page to make TA400/800 as a DHCP client.

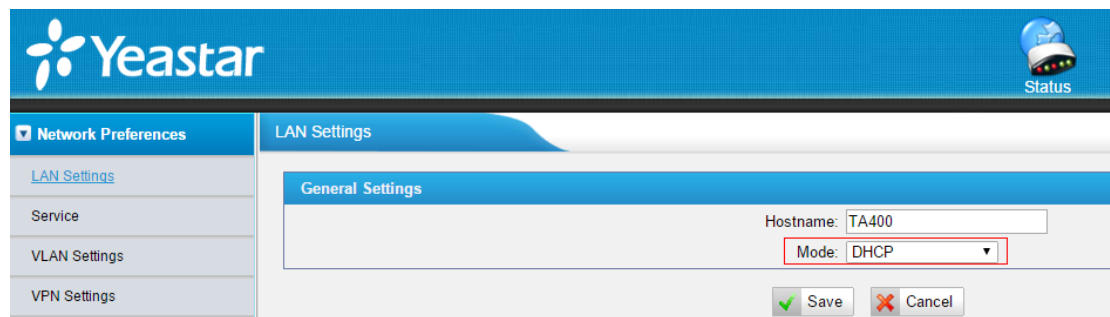


Figure 11-6 Set TA400/800 as a DHCP Client

Another way to do auto provision is to download configuration file from the configured server URL. Fill in the URL, user name, password, and set the time, TA400/800 will

get the configuration file from the server automatically and regularly.

Note: if there is no user name and password for the server, leave these fields blank.

The screenshot shows a configuration interface with two main sections: 'Server Settings' and 'Other'.
 In the 'Server Settings' section:
 - 'Server URL' is followed by an information icon and a text input field.
 - 'User Name' is followed by an information icon and a text input field.
 - 'Password' is followed by an information icon and a text input field.
 - There are two radio buttons for update frequency:
 - 'Interval of time' with a value of '180' and the unit 'Minute'.
 - 'Specified time' which is selected, with a dropdown menu showing 'Everyday', and two time dropdowns showing '00' and '00'.
 In the 'Other' section:
 - 'AES Key' is followed by an information icon and a text input field.
 - 'Always Apply' is followed by an information icon and a dropdown menu currently set to 'No'.

Figure 11-7 Server Address

- AES Key**
 If the configuration file is encrypted by AES key, you need to fill the key in this field.
- Always Apply**
 With No, it will compare the current configuration file with the last updated one, if the contents are the same no update will be applied. With Yes, it will always apply the updated configuration file.

Firmware Update

TA400/800 can be upgraded to a new firmware version via network or locally. Users could upgrade firmware via HTTP or TFTP. Please go to **System**→ **System Preferences**→ **Firmware Update** to do upgrade.

Notes:

- If “Reset configuration to Factory Defaults” is enabled, the system will be restored to factory default settings.
- When updating the firmware, please don’t turn off the power. Or the system will be damaged.
- If you are trying to upgrade through HTTP, please make sure that your TA400/800 is able to visit external network, or it cannot access Yeastar website to get the firmware file, causing the upgrade fail.

Upgrade through HTTP

On the Firmware Upgrade page, choose **HTTP URL**.

Step1. Enter the download link of the firmware file.

Note: the HTTP URL should be a **BIN** file download link.

Step2. Click “Start” to upgrade.

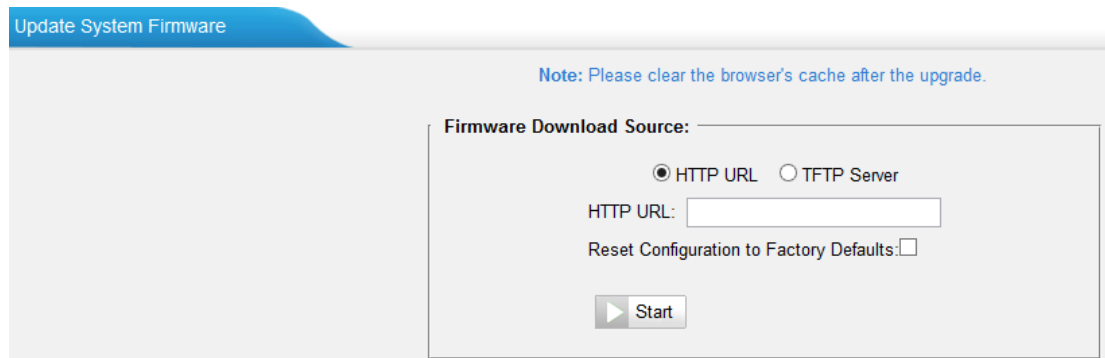


Figure 11-8 Upgrade through HTTP

Upgrade through TFTP

Step1. Download firmware file from Yeastar website.

Step2. Create a tftp Server (For example, tftpd on Windows).

1) Install tftpd32 software on computer.

Download link: http://tftpd32.jounin.net/tftpd32_download.html

2) Configure tftpd32.

On option “**Current Directory**”, click “**Browse**” button, choose the firmware file (BIN file) upgraded patch.

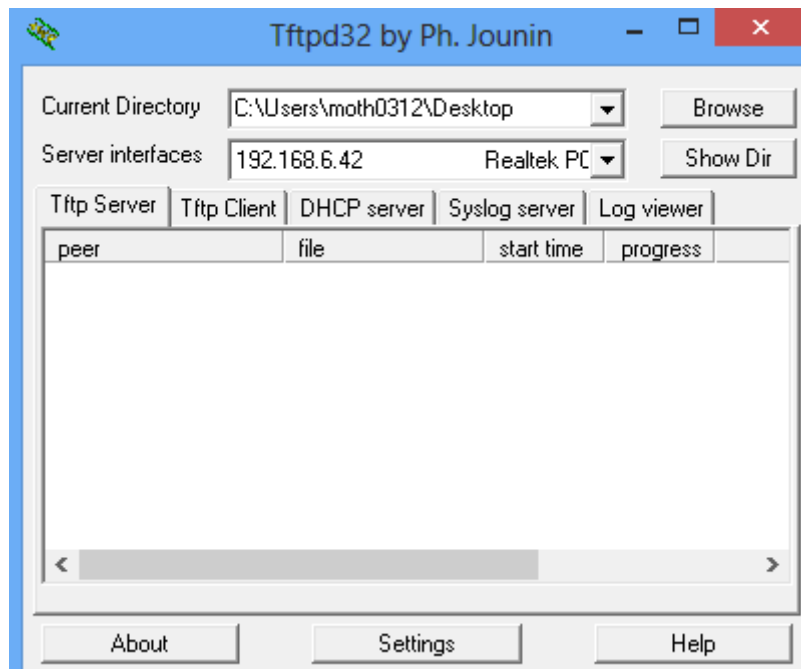


Figure 11-9 Configure Tftpd32

Step3. Logon the TA400/800's Web page and go to **System**→**System Preferences**→**Firmware Update**, choose “**TFTP Server**”.

1) TFTP Server: fill in IP address of tftpd32 server (your PC's IP address).

- 2) File Name: enter the name of firmware update. It should be a BIN file name.
- 3) Click “Start” to upgrade.

Figure 11-10 Upgrade through HTTP


Backup and Restore

TA400/800 provides Backup and Restore feature, which allows you to create a complete backup of TA400/800 configurations to a file.


Notes:

1. When you have updated the firmware version, it's not recommended to restore using an old package.
2. Backup from an earlier version cannot be restored on TA400/800 of a later version.


- **Create a New Backup**

Click  **Create a New Backup** to create a new backup.

- **Upload a Backup**

Click  **Upload a Backup** to upload a backup.

- **Restore**

To restore TA400/800 configuration data, upload the backup file to TA400/800 and click . Reboot the system to take effect.

Please note the current configurations will be **OVERWRITTEN** with the backup data.




| # | Name | Time | Options |
|---|----------------------------|-------------------------|---|
| 1 | backup_2015may9_174120.tar | Sat May 09 1:41:58 2015 |    |

Figure 11-11 Restore Backup

Reset and Reboot

Users could reset and reboot the system under **System**→ **System Preferences**→

Reset and Reboot.

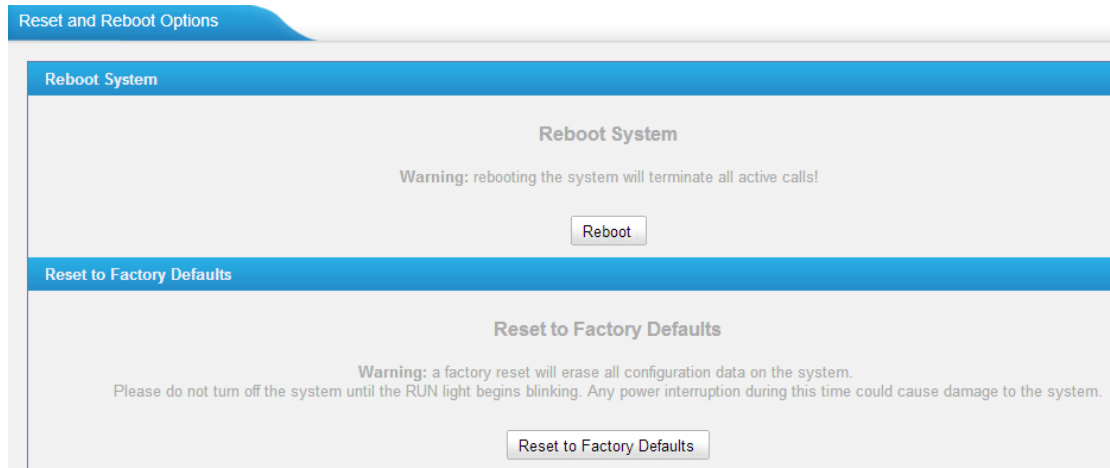


Figure 11-12 Reset and Reboot

Status

Users could check the system status on **Status**→**System Status**, where FXS Port Status, Network Status and System Info can be checked.

- FXS Port Status
- Network Status
- System Info

FXS Port Status

| Port | UP/Down | Number | Status | Voice Mail(New/Old) | Off-hook/On-hook |
|------|---------|--------|--------|---------------------|------------------|
| 1 | Up | 100 | OK | 0/0 | On Hook |
| 2 | Up | 101 | OK | 0/0 | On Hook |
| 3 | Up | 102 | OK | 0/0 | On Hook |
| 4 | Up | 103 | OK | 0/0 | On Hook |
| 5 | Up | 104 | OK | 0/0 | On Hook |

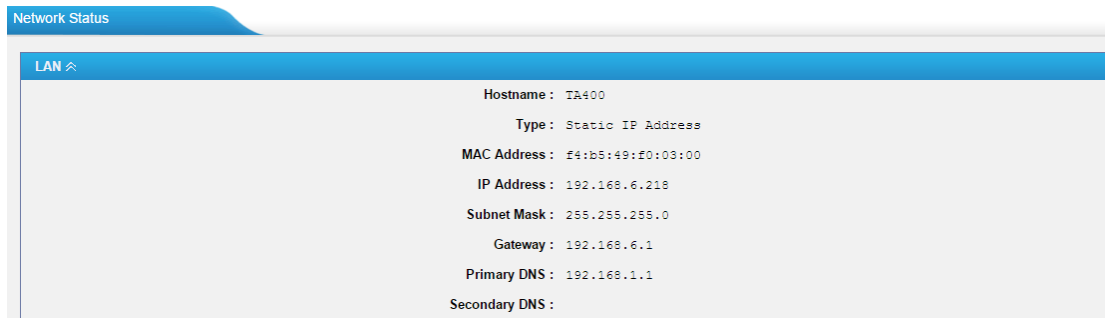
Figure 12-1 FXS Port Status

Table 12-1 Description of FXS Port Status

| Up/Down | |
|---|---|
| Up | The FXS module works well. |
| Down | The FXS module is broken. |
| Status | |
| OK | Successful registration, trunk is ready for use |
| Unreachable | The trunk is unreachable. |
| Request Send | Registering. |
| Waiting for authentication | Wrong password or user name. |
| Failed | Trunk registration failed. |
| Voice Mail (New/Old) | |
| Showing the number of unread voicemail and old voicemail. *Only when the FXS port is registered as one of the VoIP Server's accounts, can you see the voice mail status. | |
| Hook | |
| On Hook | The FXS port is idle. |
| Off Hook | The FXS port is busy. |

Network status

In this page, the IP address of LAN port will appear with their status.



The screenshot shows the 'Network Status' page with a 'LAN' section expanded. The configuration details are as follows:

| | |
|----------------|-------------------|
| Hostname: | TA400 |
| Type: | Static IP Address |
| MAC Address: | f4:b5:49:f0:03:00 |
| IP Address: | 192.168.6.218 |
| Subnet Mask: | 255.255.255.0 |
| Gateway: | 192.168.6.1 |
| Primary DNS: | 192.168.1.1 |
| Secondary DNS: | |

Figure 12-2 Network Status

If your VLAN or VPN are configured, you can check the status in this page also.

System Info

In this page, we can check the hardware/firmware version, or the disk usage of TA FXS Gateway.



The screenshot shows the 'System Info' page with the 'General' section expanded. The details are as follows:

General

- Product Type: TA400
- Hardware Version: V1.10 0000-0000
- Firmware Version: 41.19.0.16
- SN Version: A40014330049
- Uptime: 22:20:05 up 4 days, 22:18, load average: 1.26, 1.07, 1.06

Disk Usage

Notes: If there is not enough disk space on the system, the oldest call log files will be automatically deleted as necessary.

| | Used/Total (1K-blocks) | use% |
|--------|------------------------|------|
| flash: | 23060/90112 | 26% |

Memory Usage

| | Used/Total (1K-blocks) | use% |
|------|------------------------|------|
| Mem: | 76468/107796 | 70% |

Figure 12-3 System Info

Reports

Users could check the call logs, system logs on **Status**→**Reports** page, and use the packet Tool and Port Monitor Tool to capture debug logs from TA400/800.

- [Call Logs](#)
- [System Logs](#)
- [Packet Tool](#)
- [Port Monitor Tool](#)

Call Logs

The call log captures all call details, including call time, caller number, callee number, call type, call duration, etc. An administrator can search and filter call data by call date, caller/callee, trunk, duration, billing duration, status, or communication type.

The screenshot shows the 'Call Logs' interface. At the top, there is a search bar with the following filters: Start Date (04 Jun 2014), End Date (04 Jun 2014), Caller/Callee, Trunk (All), Duration, Billing Duration, Status (All), and Communication Type (All). A 'Start Searching' button is located to the right of the filters. Below the filters, there are two buttons: 'Download the recordings' and 'Delete the recordings'. To the right of these buttons, it says 'Total: 39 Show: 1-25 View: 25'. Below this is a table with the following columns: Time, Caller, Callee, Source Server/Port, Destination Server/Port, Duration, Billing Duration, Status, and Communication Type. The table contains five rows of data.

| Time | Caller | Callee | Source Server/Port | Destination Server/Port | Duration | Billing Duration | Status | Communication Type |
|---------------------|--------|---------------|--------------------|-------------------------|----------|------------------|----------|--------------------|
| 2014-06-04 22:05:08 | 304 | *741 | | | 11 | 3 | ANSWERED | Internal |
| 2014-06-04 22:02:37 | 304 | huntinggroup1 | | Port2 | 2 | 0 | ANSWERED | Internal |
| 2014-06-04 22:02:34 | 304 | 300 | SOHO | | 80 | 80 | ANSWERED | Inbound |
| 2014-06-04 22:02:28 | 304 | 300 | Port3 | SOHO | 86 | 80 | ANSWERED | Outbound |
| 2014-06-04 22:01:59 | 304 | 300 | Port3 | SOHO | 5 | 0 | FAILED | Outbound |

Figure 13-1 Call Logs

System Logs

You can download and delete the system logs of TA400/800.

- **Enable Hardware Log**
Save the information of hardware; (up to 4 log files)
- **Enable Normal Log**
Save the prompt information; (up to 16 log files)
- **Enable Web Log**
Save the history of web operations (up to 2 log files)
- **Enable Debug Log**
Save debug information (up to 2 log files)

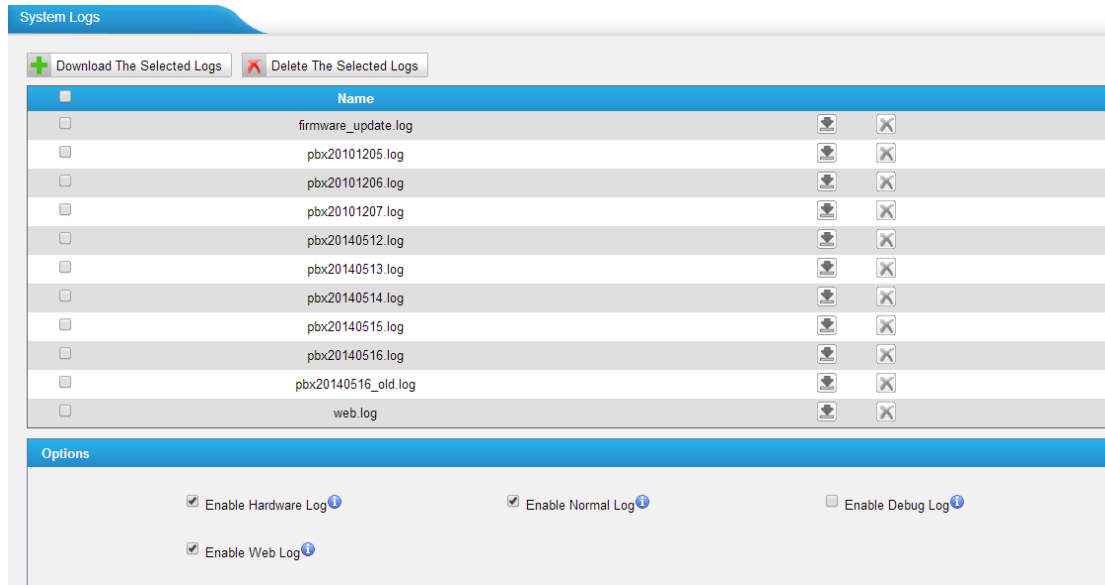


Figure 13-2 System Logs

Packet Tool

This feature is used to capture packets for technician. Integrate packet capture tool “Wireshark” in TA400/800. Users also could specify the destination IP address and port to get the packets.

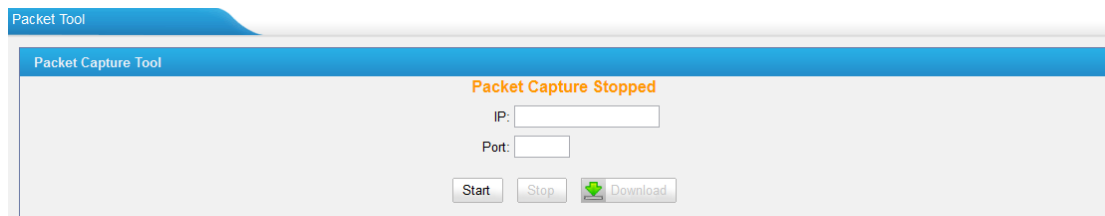


Figure 13-3 Packet Tool

- **IP**
Specify the destination IP address to get the packets.
- **Port**
Specify the destination Port to get the packets.

Port Monitor Tool

This tool is used to debug a FXS port. Select a FXS port and click “Start” to monitor the FXS port, stop monitoring by clicking “Stop” button.



Figure 13-4 Port Monitor Tool