



Welltech 2540 FXO Gateway User Guide

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Welltech 2540

CH1 Introduction

2540 Telephony Gateway

The Welltech 2540 is a 4 ports FXO (WellGate 2540) VoIP gateway which includes 1-WAN/1-LAN (management port) 10/100 base-T network environment. Field-proven quality of Voice communication and Fax transmission over IP broadband access network to makes WellGate 2540 to be an excellent solution for various VoIP applications.

1-1 Physical Interface

- Ethernet port (RJ-45, 10/100 base-T)
 - 1-WAN port, for connect to router, ADSL modem (ATU-R), or switch hub directly.
 - 1-LAN port, for PC, management or other network devices connecting.
- Telephony port (RJ-11)
 - 24-FXO ports, to connect to PSTN lines
- Console port (RS232, rate: 115200)
- AC power Jack
- Status indicated LED
Indicates Power, Ethernet, Line, SIP and system status

1-2 IP Network connection

IPv4 (RFC 791)/IPv6 (RFC 2460)
IPv6 Auto Configuration (RFC 4862)
MAC Address (IEEE 802.3)
Static IP
DHCP Client (RFC 2131)
PPPoE
DNS Client

TCP/UDP (RFC 793/768)
 RTP/RTCP (RFC 1889/1890)
 IPV4 ICMP (RFC 792)/IPV6 ICMP (RFC 4443)
 TFTP Client
 VOIP VLAN Support (802.1q/802.1p)
 HTTP/HTTPS Server
 QoS Support
 Support IPV4 only, IPV6 only or dual stack mode

1-3 Environmental

- Actual Dimension: 35 × 242 × 160 mm (Desktop)
- Weight: 0.935kg (unit without packing)
- Operating Temp. & Humidity
 - Temp.: 0°C ~ 45°C (32°F ~ 113°F)
 - Humidity: 10% ~ 85% relative humidity, non-condensing
- Storage Temp. & Humidity
 - Temp.: 0°C ~ 55°C (32°F ~ 131°F)
 - Humidity: 10% ~ 95% relative humidity, non-condensing
- AC to DC Power Adaptor:
 - INPUT: AC100V-240V, 50/60Hz
 - OUTPUT: DC 12V, 1.5A
- Regulatory Compliance: FCC (Part 15, Class B) & CE

1-4 Front Panel: LED Indicators

WellGate 2540



Figure 1-4-1 front panel

LED	Description
Power	When the power adapter is connected, the LED will light up green.
Status	When system is startup successfully, the LED will light up

	green.
Proxy	When the gateway is registered successfully to a Proxy, this will light up green.
WAN	This will light up green when the gateway's WAN port is physically connected to the public internet. When data is transmitted through this port, it will flash green. The default IP of WAN port is 10.1.1.3.
LAN	This will light up green when the gateway's LAN port is physically connected to a local network (Refer to Rear Panel section in page number for location of LAN port). When data is transmitted through this port, it will flash green. The default IP of LAN port is 192.168.123.123.
Line1 ~Line4	The status LED for FXO port 1-4, this will light up amber orange when the connected phone is engaged in a conversation. It will flash amber orange when there is an incoming call.

1-5 Rear Panel: LED Indicators

WellGate 2540

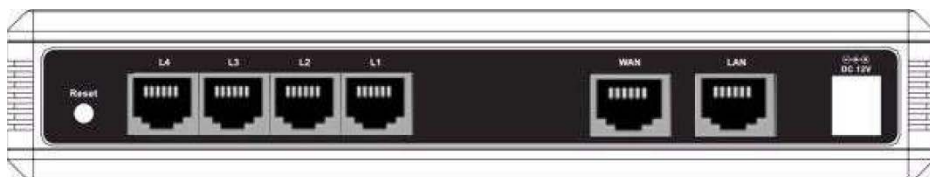


Figure 1-4-1 rear panel for FXO

Item	Description
Reset	Press and hold over 5 seconds to reload factory default setting, this will erase all existing settings configured on this gateway.
L1 - L4	The status LED for FXO port 1-4. When there is no PSTN connected, this LED will become blinking. When PSTN is connected and no talking, the LED will be off. When a line is using, the LED will become light up. (WellGate 2540 only)
LAN	10/100 Base-T RJ-45 socket for LAN port, connects to PC for management purpose.

WAN	10/100 Base-T RJ-45 socket for WAN port, connects to wide area network.
AC100V~240V	The power socket, input AC 100V~240V, output DC 12V, 6A.

1-6 QUICK SETUP

Note:

Please use Windows XP IE 6.0 web browser or above version to configure FXO gateway webpage setting. Welltech products don't support other Web Browser such as FireFox to configure.

Login :

Setp1: Setup the administrative PC's IP address to be same as Welltech 2540 and connect the Ethernet cable into WAN or LAN port. Start IE6.0 (or later version) to navigate Welltech 2540 web management system by typing the default URL which is <http://192.168.123.123> (through LAN port) or <http://10.1.1.3> (through WAN port). The screen will display User Name and Password (the default user id is **root** and user password is **root**). (See figure 1-6-1 web access)



Figure 1-6-1 web access

Step 2: After login, the screen shows the Home page of Welltech 2540. (See figure 1-6-2 Network configure-1)

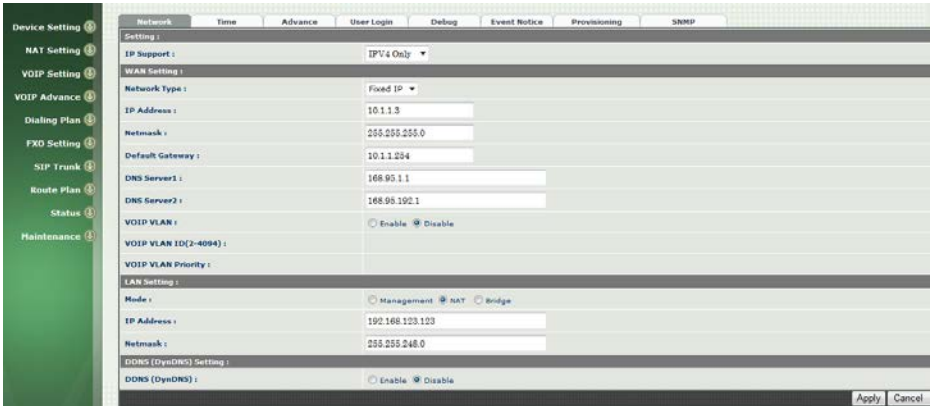


Figure 1-6-2 Network configure-1

Change Default IP Network:

Step 3: After successfully logon to the system, we need to change the network configuration. Click **Device Setting > Network** to setup the service network interface (WAN) parameters. Enter the deserved IP address, netmask and default gateway or selected to "DHCP" or "PPPOE". Apply the change by clicking **Apply** button as fig (See Figure 1-6-3 Network configure-2).

Note: If Gateway WAN port are setting in the 10.x.x.x segment, please make sure that you also change the LAN port to other segment such as 192.168.x.x

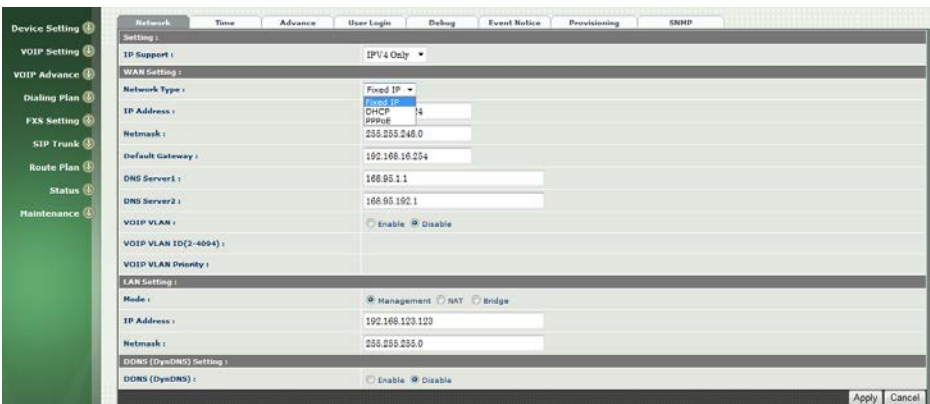


Figure 1-6-3 Network configure-2

Change Default Time setting:

Step 4: When re-logout to the new IP address, the next is to setup the system time zone. Click **Device Setting > Time** to setup the system. Enter the current SNTP server, time zone and daylight saving parameters. Apply the change by clicking **Apply** button. (See figure 1-6-4 Time setting)

The screenshot shows the 'Time' configuration page. The left sidebar lists 'Device Setting' with sub-items: NAT Setting, VOIP Setting, VOIP Advance, Dialing Plan, FXD Setting, SIP Trunk, Route Plan, Status, and Maintenance. The main content area has tabs for Network, Time, Advance, User Login, Debug, Event Notice, Provisioning, and SNMP. The 'Time' tab is active, showing the following settings:

- Current Time: 2010/01/25 09:27:54
- NTP Time Server: 168.95.195.12
- NTP Refresh Interval(sec): 43200
- Time Zone: GMT
- Daylight Saving: Yes No
- Daylight Bias: +05:00
- Daylight Start: Month: [dropdown], Week Day: Sun, Apply Week: 1, Hour: 00
- Standard Start: Month: [dropdown], Week Day: Sun, Apply Week: 1, Hour: 00

Buttons for 'Apply' and 'Cancel' are at the bottom right.

Figure 1-6-4 Time setting

Modify SIP Account Parameter:

Step 5: The next step is to add a SIP trunk for VOIP calling. For WellGate 2540, it is necessary for VOIP calling. Click **SIP trunk and new to** create the required sip trunk. Enter the trunk ID to 1 and input those SIP parameters. Apply the change by clicking **Apply** button. (See Figure 1-6-5 SIP Trunk)

The screenshot shows the 'SIP Trunk' configuration table. The left sidebar is the same as in Figure 1-6-4, with 'SIP Trunk' highlighted. The table has the following columns: Trunk ID, Register Type, TEL No, Proxy Server, Proxy Server Port, Outbound Proxy, and Outbound Server Port.

Trunk ID	Register Type	TEL No	Proxy Server	Proxy Server Port	Outbound Proxy	Outbound Server Port
1	Register	254001	10.1.1.2	5060		5060

Buttons for 'New', 'Total Record: 1', 'Total Page: 1', and 'Page 1' are at the bottom right.

Figure 1-6-5 SIP Trunk

Soft Reset Welltech 2540:

Step 6: After modify basic setting. It is required to reset Welltech 2540. Click **Maintenance > Maintenance > Soft-Reset or Reboot** to take effect. Apply the change by clicking **Apply** button. (See Figure 1-6-6 Quick-Reset)



Figure 1-6-6 Quick-Reset

Check Welltech 2540 Registered Status:

Step 7: After soft-reset or reboot.

>Click **Status > SIP Trunk Status** to check whether registered or not. (See figure 1-6-7 SIP trunk status)



Figure 1-6-7 SIP trunk status

Through the above settings, the Welltech 2540 should be able to do the following:

For FXO (Welltech 2540):

1. For PSTN incoming call, the caller will hear a dial tone. Then the caller can dial a VOIP number and it will use the setup SIP Trunk 1 to make SIP call out. (2 stage dialing)
2. For VOIP incoming call, Welltech 2540 will off hook a FXO port and dial to PSTN.

CH2 Device Settings

From this setting category, all devices related parameters can be found here.

2-1 Network Configuration

> Network

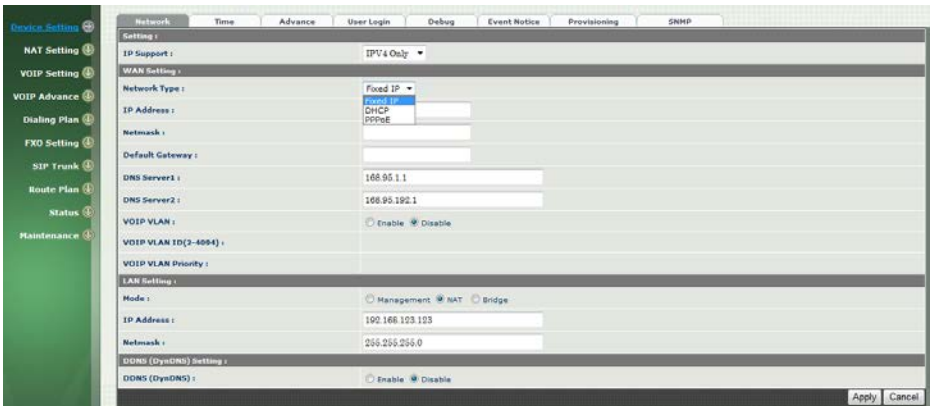


Figure 2-1 network setting

Parameter Description:

Setting:

- **IP Support:** IP stack to be supported (IPV6 and IPV4 or IPV6 or IPV4 only)

WAN Setting:

- **Network Type:** support "Fixed IP"; "DHCP"; "PPPOE"
- **IP Address:** IP address
- **Default Gateway:** Default gateway
- **DHCP Tag (option 60):** input Vendor class identifier or not.
- **DHCP Tag (option 61):** input Client identifier or not.
- **DNS Server1:** Primary DNS Server IP network
- **DNS Server2:** Secondary DNS Server IP network
- **VOIP VLAN:** Enable VOIP VLAN or not. When enable VOIP VLAN, the WAN port can be only accessed by VLAN. If it is required to manage the

WellGate 2540, Administrator can use LAN port instead.

- **VOIP VLAN ID(2-4096):** VLAN ID Used

Note: the default WAN IP address is 10.1.1.3.

LAN Setting:

- **Management mode:** This LAN port is used for management purpose, not used for register or routing.
- **NAT mode:** DHCP function on the LAN port. The LAN ports will function as a DHCP server, network devices connected to them will be issued with IP addresses. (On the left item will add a NAT setting, the information please refer NAT setting)
- **IP Address:** IP address (please set to 192.168.x.x if your WAN port is using 10.x.x.x IP segment, the IP address must different as WAN IP).
- **Netmask:** IP network mask
- **Bridge mode:** At this mode, both WAN and LAN ports are configured to Switch/Hub features. LAN port access to WAN port directly.

Note: default LAN IP address is 192.168.123.123

DDNS (DynDNS) Setting:

- **DDNS (DynDNS):** enable or disable dynamic DNS feature.
- **Domain Name:** input your Domain Name
- **User Name:** input your user name
- **Password:** input your password

2-2 Device Time Setting

Welltech 2540 support SNTP with time zone and daylight saving.

Device Setting > Time

Network	Time	Advance	User Login	Debug	Event Notice	Provisioning	SNMP
Current Time:		2011/12/19 09:45:34					
NTP Time Server:		188.85.195.12					
NTP Refresh Interval(sec):		43200					
Time Zone:		GMT+08:00					
Daylight Saving:		<input checked="" type="radio"/> Yes <input type="radio"/> No					
Daylight Bias:		+00:00					
Daylight Start:		Month:	08	Week Day:	Mon	Hour:	00
Apply Week:		2					
Standard Start:		Month:	09	Week Day:	Sun	Hour:	00
Apply Week:		3					
<input type="button" value="Apply"/> <input type="button" value="Cancel"/>							

Figure 2-2 Time setting

Parameter Description:

- **Current Time:** display time of now (display only)
- **NTP Time Server:** SNTP time server
- **NTP Refresh Interval(sec):** The frequency to sync NTP server in seconds
- **Time Zone:** The time-zone Welltech 2540 is located.
 - Standard: Use a predefined standard time zone
 - Customize: Use a user defined time zone
- **Daylight Saving:** Auto adjust daylight saving timer or not
- **Daylight Bias:** The offset added to the Bias when the time zone is in daylight saving time
- **Daylight Start:** The date that a time zone enters daylight time
 - Month: 01 to 12
 - Week Day: Sunday to Saturday
 - Apply Week (Day:01 to 05, Specifies the occurrence of day in the month; 01 = First occurrence of day, 02 = Second occurrence of day, ...and 05 = Last occurrence of day)
 - Hour: 00 to 23
- **Standard Start:** The date that a time zone enters daylight time
 - Month: 01 to 12
 - Week Day: Sunday to Saturday
 - Apply Week (Day:01 to 05, Specifies the occurrence of day in the month; 01 = First occurrence of day, 02 = Second occurrence of day, ...and 05 = Last occurrence of day)
 - Hour: 00 to 23

2-3 Device Advance Setting

> Advance



Figure 2-3 Advance setting

Parameter Description:

- **HTTP Service:** The Administrator Web service port (the default is 80)
- **HTTPS Service:** The https web service port (the default is 443)
- **Telnet Service:** The telnet service port (the default is 23)
- **HTTP/HTTPS Service access on WAN:** When click the disable option; The WEB service will be rejected on WAN port, so please be careful with this function. If you wanted to enable WAN port again, you need to access this device from its LAN port to connect to WEB pages and enable WAN port.

2-4 User Login Setting

Three level of users can be used, administrator, supervisor, user. Each level of users will have different predefined access level.

>User Login

Section	User ID	Password	Confirm Password	Language
Administrator	root	****	****	English
Supervisor	admin	****	****	English
User	user	****	****	English

Figure 2-4 user login setting

Parameter Description:

- **Administrator:** The administrator level user which has full access of Welltech 2540.
- **Supervisor:** The supervisor level user which has limited administrative access right.
- **User:** The user access right which only allows to setting some user related features.
- **User ID:** Login User ID
- **Password:** Login Password
- **Confirm Password:** Confirm new password again
- **Language:** The web page language used when the account login. To add a customized local language, please contact Welltech.

2-5 Debug Settings

Welltech 2540 provides the real time debug to syslog or through telnet interface. It generates the debug information system based on debug level and modules. Since the generating debug will consume system resource, it is recommended to turn on only for necessary and under Welltech FAE's instruction.

Debug

The screenshot shows the 'Debug' configuration page in the Welltech 2540 web interface. The page is organized into several sections:

- phoneMgr:** Includes 'Debug Module' (checkboxes for Device Control, Call Control, DB, Verbose) and 'Debug Level' (dropdown set to Emergency).
- SipMgr:** Includes 'Debug Module' (checkboxes for Register, Call, Sip Message, Other) and 'Debug Level' (dropdown set to Emergency).
- emesck:** Includes 'Debug Level' (dropdown set to Emergency).
- SYSLOG:** Includes radio buttons for 'Enable' and 'Disable' (checked), and a 'Check for start from Any Time' checkbox.
- DSP Debug:** Includes radio buttons for 'Enable' and 'Disable' (checked).

Additional fields include Syslog Start (YYYY/MM/DD HH:MM), Syslog Stop (YYYY/MM/DD HH:MM), Syslog Server, Syslog Port, DSP Capture Server, and DSP Capture Port. The bottom right corner has 'Apply' and 'Cancel' buttons.

Figure 2-5 Debug setting

Parameter Description:

- **SYSLOG:** Enable or disable to send system information to SYSLOG server or not
- **Check for start from Any Time:** Always Send: Always send syslog or only during a specified time range.
- **Syslog Start (YYYY/MM/DD HH:MM):** Always Send: Always send syslog or only during a specified time range.
- **Syslog Stop (YYYY/MM/DD HH:MM):** The syslog stop sending time.
- **Syslog Server:** Syslog server IP address
- **Syslog Port:** syslog server service port (default is 514)
- **DSP Debug:** Enable or disable to send DSP information to capture log
- **DSP Capture server:** Syslog capture server IP address
- **DSP Capture port:** syslog capture server service port (default is 50000)

2-6 Event Notice

Welltech 2540 can send Syslog Event Notice when it had the following cases:

1. Register Failure or re-registered
2. FXO RJ-11 cable was plugged or unplug
3. Ethernet reconnected
4. System started



Figure 2-6 Event notice setting

- **Syslog Notice:** Enable or disable to send system event to SYSLOG server or not
- **Syslog Server:** Syslog server IP address
- **Syslog Port:** syslog server service port (default is 514)

2-7 Auto Provisioning

The Welltech 2540 can be provisioned by WellEMS 9510 for large deployment. Please contact Welltech for availabilities.

>Provisioning

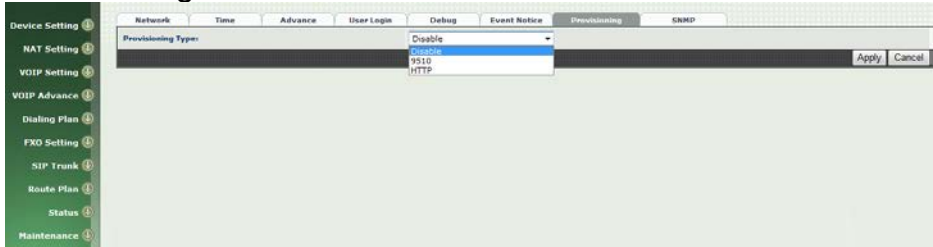


Figure 2-7-1 Provisioning

Select 9510:

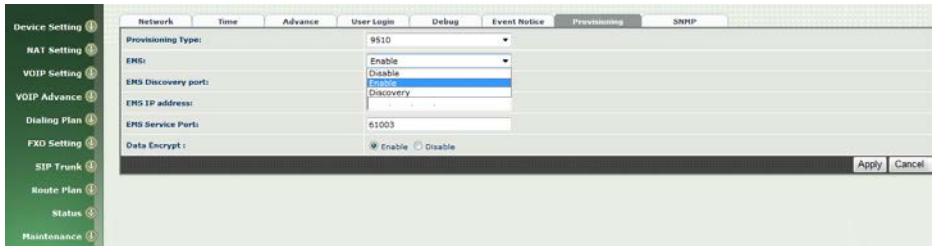


Figure 2-7-2 Provisioning type of 9510

Parameter Description:

(This function is not available yet for WellEMS 9510)

- **EMS:** Enable auto provisioning service by WellEMS 9510 or not.
 - **Enable:** Enable the service and use manual configured EMS server parameters.
 - **Disable:** Disable the auto provisioning service.
 - **Discovery:** To automatically discover the EMS server or not. By using this mode, WellEMS 9510 need to be the same IP network in order to make the IP broadcasting work.
- **EMS Discovery Port:** WellEMS 9510 service auto discovery broadcasting port (default is 61005).
- **EMS summary refresh interval:** How long the Welltech 2540 will report its summary status to WellEMS 9510 in seconds.

- **EMS IP address:** The WellEEMS 9510 server IP address
- **EMS Server Port:** The WellEEMS 9510 Server port
- **Data Encrypt:**

Disable: disable encryption function.

Welltech encryption: Enable Welltech proprietary encryption for SIP signaling and RTP. It is required a Welltech SIP proxy server (WS6500 or SIPPBX 6200) to work with this feature. When enable it, you can hide your VOIP traffic from ISP's monitor.

External encryption: for custom encryption, it is valid now, if you want add the function to mach your proxy, please contact with Welltech's sales.

Select Http:

This feature is for feature usage only.

The screenshot shows a web-based configuration interface for provisioning. On the left is a vertical navigation menu with options: Device Setting, NAT Setting, VOIP Setting, VOIP Advance, Dialing Plan, FXO Setting, SIP Trunk, Route Plan, Status, and Maintenance. The main content area has tabs for Network, Time, Advance, User Login, Debug, Event Notice, Provisioning (selected), and SNMP. Under the Provisioning tab, the 'Provisioning Type' is set to 'HTTP'. Below this, there are input fields for 'HTTP Config URL' (containing 'http://192.168.17.30/24xx_provision/'), 'Refresh Interval (minute)' (set to '60'), 'User ID' (set to 'welltech'), and 'Password' (masked with asterisks). 'Apply' and 'Cancel' buttons are at the bottom right.

Figure 2-7-3 Provisioning type of Http

- **Http config URL:** internal used only
- **Refresh interval(minute):** interval to check whether have a new configuration/firmware or not in minutes
- **User ID:** specify the login id for http authentication
- **Password:** specify the password for http authentication

2-8 SNMP

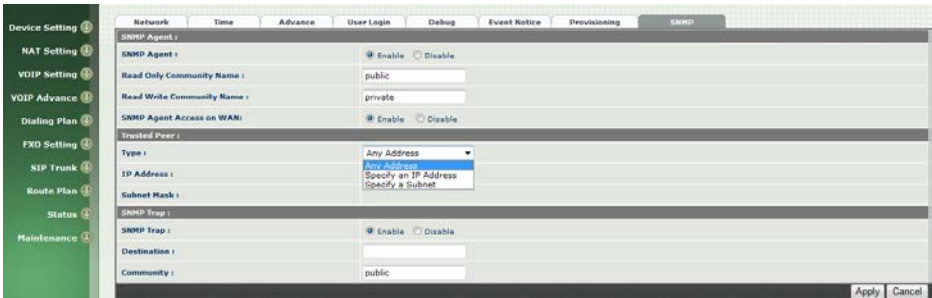


Figure 2-8 SNMP

SNMP Agent:

- **SNMP Agent:** Enable SNMP or not.
- **Read Only Community Name:** The community name to read through SNMP protocol
- **Read Write Community Name:** The community name to read and write through SNMP protocol.
- **SNMP Agent Access on WAN:** Enable SNMP to be accessed through WAN port or not.

Trusted Peer:

- Type:
 - ◆ **Any Address:** Any address can retrieve the SNMP information.
 - ◆ **Specify an IP Address:** Only the IP address listed can retrieve the SNMP information. Normally, it will be the SNMP manager IP address.
 - ◆ **Specify a Subnet:** Only the network specified can retrieve the SNMP information.
- **IP address:** The IP address for a trusted peer
- **Subnet Mask:** The network mask for a trusted peer

SNMP Trap:

- **SNMP Trap:** Enable SNMP trap or not
- **Destination:** The IP address for SNMP manager to receive the SNMP trap
- **Community:** The communicate name for sending the SNMP trap

CH3 NAT Setting

The WellGate 2540 can support NAT, 2 ethernet leg (gw) or bridge mode. Here are the settings for NAT related service.

3-1 DHCP Ser. (DHCP server)

DHCP Ser.	UPnP	Bandwidth	URL Filter	IP Filter	MAC Filter	App Filter	Port Filter	Port Fwd.
DHCP Server :	<input checked="" type="radio"/> Enable <input type="radio"/> Disable							
DHCP Client Range (Start IP):		192.168.123.1						
DHCP Client Range (End IP):		192.168.123.100						
DHCP Router Opt.:		192.168.123.123						
DHCP Subnet Opt.:		255.255.255.0						
DHCP DNS1 Opt.:		168.95.1.1						
DHCP DNS2 Opt.:		168.95.192.1						

Figure 3-1 DHCP server

- **DHCP Server:** Enable DHCP server or not.
- **Client Range Start IP:** specify DHCP client lease start IP
- **Client Range End IP:** specify DHCP client lease end IP
- **Default Gateway:** specify the default gateway
- **Submask:** specify the submask.
- **DNS Server 1:** specify the DNS server
- **DNS Server 2:** specify the DNS server

3-2 UPNP (universal plug and play server)



Figure 3-2 UPnP

- **UPNP IGD:** Enable UPNP server or not.

3-3 Bandwidth (Bandwidth Control)

By using bandwidth control feature, the user can manage the traffic based on their needs.



Figure 3-3-1 Bandwidth control

Bandwidth Control:

- **Bandwidth Control:** enable bandwidth control or not.
- **Download Bandwidth:** specify total bandwidth for download (unit: kbps). 0 indicates no limitation.
- **Upload Bandwidth:** specify total bandwidth for upload (unit: kbps). 0 indicates no limitation.

Maximum Bandwidth and Reserved Bandwidth:

- **Setup Method:** bandwidth control method, percentage or specify the required bandwidth
- percentage : total bandwidth
 - ◆ **priority 1:** highest priority percentage
 - ◆ **priority 2:** Normal priority percentage
 - ◆ **priority 3:** low priority percentage

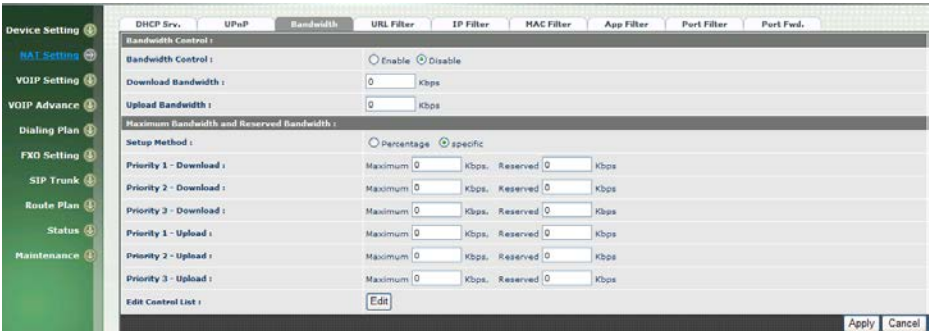


Figure 3-3-2 Bandwidth control

- specific :
 - ◆ **priority 1 – Download:** highest priority download bandwidth
 - ◆ **priority 2 – Download:** normal priority download bandwidth
 - ◆ **priority 3 – Download:** low priority download bandwidth
 - ◆ **priority 1 – Upload:** highest priority upload bandwidth
 - ◆ **priority 2 – Upload:** normal priority upload bandwidth
 - ◆ **priority 3 – Upload:** low priority upload bandwidth



Figure 3-3-3 Edit control list

In order to set which target is belonged to which priority, the following is the setting method for target's priority.

IP Target

The screenshot shows the 'Create Control List' configuration page. On the left is a navigation menu with categories: Device Setting, NAT Setting, VOIP Setting, VOIP Advance, Dialing Plan, FXO Setting, SIP Trunk, Route Plan, Status, and Maintenance. The main area contains the following fields:

- Priority: 1
- Type: IP
- Configure Type: Unique (selected), IP Range
- IP Address: none

Buttons: Apply, Cancel, Back

Figure 3-3-4 IP Target 1

The screenshot shows the 'Create Control List' configuration page. On the left is a navigation menu with categories: Device Setting, NAT Setting, VOIP Setting, VOIP Advance, Dialing Plan, FXO Setting, SIP Trunk, Route Plan, Status, and Maintenance. The main area contains the following fields:

- Priority: 1
- Type: IP
- Configure Type: Unique, IP Range (selected)
- Start IP: none
- End IP: none

Buttons: Apply, Cancel, Back

Figure 3-3-5 IP Target 2

- **Priority:** Priority value for the target
- **Type:** The target type is set to IP
- **Configure Type:** unique IP or a range of IP address
 - Unique:
 - ◆ **IP Address:** the IP address to be set
 - IP Range:
 - ◆ **Start IP:** The starting IP for a range
 - ◆ **End IP:** The stopping IP for a range

Port Target

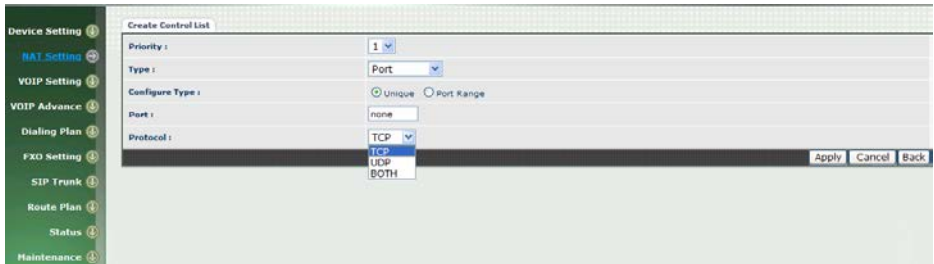


Figure 3-3-6 Port Target

- **Priority:** Priority value for the target
- **Type:** The target type is set to port number
- **Configure Type:** unique port number or a range of port number
- Unique:
 - ◆ **Port:** the port number to be added
 - ◆ **Protocol:** protocol for the port
- Port Range:
 - ◆ **Start port:** the starting port number
 - ◆ **End port:** the stop port number
 - ◆ **Protocol:** protocol for the port range

Application Target

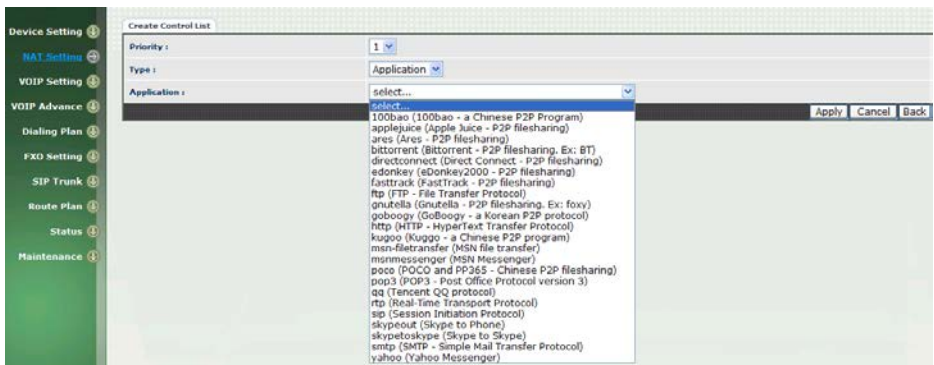


Figure 3-3-7 Application Target

- **Priority:** Priority value for the target
- **Type:** Application
- **Application:** the list for the application

DSCP target



Figure 3-3-8 DSCP Target

- **Priority:** Priority value for the target
- **Type:** DSCP value
- **DSCP:** The DSCP will be mapped to the priority

The WellGate 2540 support firewall features as below.

3-4 URL Filter



Figure 3-4 URL Filter

- **URL Filter:** the specified url will be blocked

3-5 IP Filter



Figure 3-5 IP Filter

- **IP Filter:** The specified IP address to be blocked
- **Local IP address:** The LAN side IP address to be forwarded
- **Protocol:** TCP, UDP or both are used for port forward

3-6 MAC Filter



Figure 3-6 MAC Filter

- **MAC Filter:** The MAC address to be blocked

3-7 APP Filter



Figure 3-7 App Filter

- **APP Filter:** application to be blocked

3-8 Port Filter



Figure 3-8 Port Filter

- **Port Filter:** enable port Filter or not.
- **Port Range:** Starting and stopping port to be forward. If you are using only 1 port, please set the starting equal to stopping port.
- **Protocol:** TCP, UDP or both are used for port blocked.

3-9 Port Fwd

The WellGate 2540 support port forward feature as below

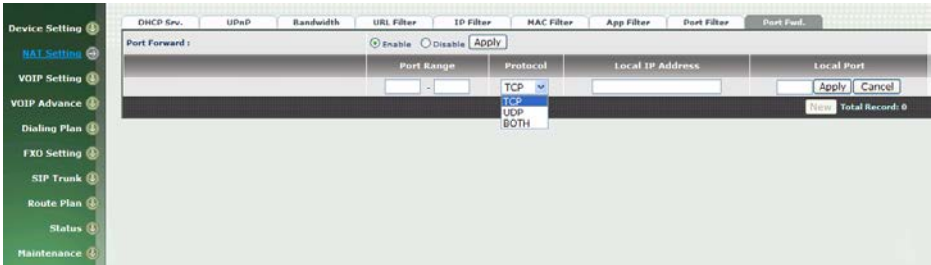


Figure 3-9 Port Fwd

- **Port Fwd:** enable port forward feature or not
- **Port Range:** Starting and stopping port to be forward. If you are using only 1 port, please set the starting equal to stopping port.
- **Protocol:** TCP, UDP or both are used for port forward
- **Local IP address:** The LAN side IP address to be forwarded
- **Local Port:** The LAN side port to be forwarded. If you are using the port range, this port indicates the starting port.

VOIP Parameters Setting

SIP Parameters:

CH4 VOIP Setting

4-1 SIP



Figure 4-1 SIP setting

Parameter Description:

- **Session Timer:** Enable session timer or not (RFC 4028)
- **Session Expires (sec):** This is the setting of initial session timer expires time according to RFC4028 - Session Timers in the Session Initiation Protocol.
- **Min SE (sec):** The minimum session timer allowed when receiving a call with session timer value according to RFC 4028.
- **PRACK:** Enable provisioning ACK or not (RFC 3262)
 - None: Disable PARCK
 - Supported: When select this mode, 100rel will be added to the support list. It indicates Welltech 2540 can support the PRACK but not mandatory.
 - Require: PRACK is mandatory required.
- **SIP Local Port:** The SIP local service port (default is 8080)
- **SIP Qos Type:** Quality of Service Type for SIP signaling
 - None: Not using QOS Tag and not enables QOS.
 - DiffServ: Differentiated Services Value. Input DSCP value 0-63 for DSCP
 - TOS: Type of Service which include IP precedence value and TOS.
- **Accept Proxy Only:** Only accept the call coming from the SIP proxy. Not accept peer to peer call at this mode

4-2 Audio



Figure 4-2 Audio setting

- **Codec 1 ~5:** The preferred codec priority
- **G.711u Payload Size:** G.711 u-Law payload size
- **G.711a Payload Size:** G.711 A-law payload size
- **G.729 Payload Size:** G.729A payload size
- **G.723.1 Payload Size:** G.723.1 payload size
- **Bit Rate:** G.723.1 bit rate used
 - ◆ 5.3K bit rate is used
 - ◆ 6.3K bit rate is used
- **Codec Priority:** Selection order to match the remote SDP for codec selection.
 - Local SDP Order:** Use local SDP order to match codec
 - Remote SDP Order:** Use Remote SDP order to match codec
- **DTMF Relay:**
 - In-Band DTMF:** use inband DTMF instead of out of band.
 - RFC 2833(fall back to SIP-INFO):** Use RFC 2833 if the SDP negotiation could be done. Or use SIP INFO for DTMF relay.
 - SIP INFO:** Use SIP-INFO DTMF relay
 - RFC 2833(fall back to Inband):** Use RFC 2833 if the SDP negotiation could be done. Or use inband DTMF transmission.
- **Silence Suppression:**

Enable: Start the voice activity (silence) detection when detect silence for 60 seconds, it will hang up the call. **(For FXO use)**

Disable: Send silence packet as normal voice packet (no silence detection)

- **RTP Basic Port:** The RTP starting port. Each channel will be add additional 10. For example, the RTP basic port is 16384, thus call 1 will use 16384 while call 2 will use 16394 etc.
- **RTP Qos Type:** IP QoS tag for RTP stream
 - DiffServ:** The differentiated service QoS tag will be used. Input DSCP value 0-63 for DSCP.
 - TOS:** Type of Service which include IP precedence value and TOS.

4-3 Tone

The setting page is used to setup the tone to be generated or detected. The detected tone is the Disconnect 1 & 2 (**for FXO use**) and the others are for generating (**when FXS received the “bye” from IP side or waiting time out by analog phone which keeps handset pick up, it will send busy tone to analog phone**). The disconnect tone is very important for PSTN status supervision.

Tone \ Setting	Signal Type	Freq 1 (0-300-1500Hz)	Freq 2 (0-300-1500Hz)	Level 1 (0-6.3dB)	Level 2 (0-6.3dB)	On 1 (0-102.50ms)	Off 1 (0-102.50ms)	On 2 (0-102.50ms)	Off 2 (0-102.50ms)	Deviation
Dial	Continuous	350	440	13	13	500	0	0	0	10
Stutter Dial	Cadence	380	440	13	13	1000	100	0	0	10
Ring Back	Cadence	440	480	13	13	1000	2000	0	0	10
Busy	Cadence	480	620	13	13	500	500	0	0	10
Call Waiting	Cadence	350	440	13	13	250	250	250	0	10
ROH	Continuous	1400	1750	13	13	10000	0	0	0	10
Warning	Cadence	900	0	13	13	500	0	0	0	10
Holding	Cadence	900	0	13	13	800	800	0	0	10
Disconnect 1	Cadence	480	620	13	13	500	500	0	0	10
Disconnect 2	Cadence	480	620	13	13	250	250	0	0	10

Figure 4-3 Tone setting

Please use Country Template to select the country profile which will be applied. Click Use to load those country tone parameters to system and change if necessary. For those countries are not showed in the list, please select a closed country and edit to match your country. You can send an email with the tone definition to Welltech if you would like to put your country into the list.

Note: about how to record disconnect tone and analyzer it. Please link to download and refer it.

http://www.welltech.com/support/voip2/SIP%20series/FXSO%20series/25xx/FXO_Tone_Analyzer/FXO%20Tone%20Analyzer_EN.zip

4-4 NAT Traversal

The WellGate 2540 support the following NAT traversal methods

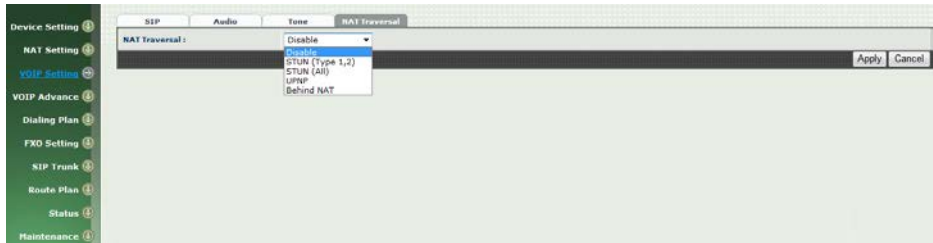


Figure 4-4 NAT Traversal

- NAT Traversal:
 - ◆ **Disable:** Disable NAT traversal features
 - ◆ **STUN (Type 1,2):** Enable STUN for NAT traversal. Since STUN can be used only for type 1 and type 2 NAT server, it is recommended to use this option. When STUN client detect the used NAT is type 3 NAT, it will stop the STUN feature.
 - ◇ **STUN Server:** STUN Server IP address
 - ◆ **STUN (All):** No matter which NAT type server are used, STUN is always to be used for NAT traversal.
 - ◇ **STUN Server:** STUN Server IP address
 - ◆ **UPNP:** Enable UPnP client for NAT traversal. Please note that the IP sharing box need support uPnP feature.
 - ◆ **Behind NAT:** Use DMZ for NAT traversal
 - ◇ **IP Sharing Address:** public IP sharing address. You need to specify the port mapping or DMZ for all required port.

CH5 VOIP Advance

5-1 SIP

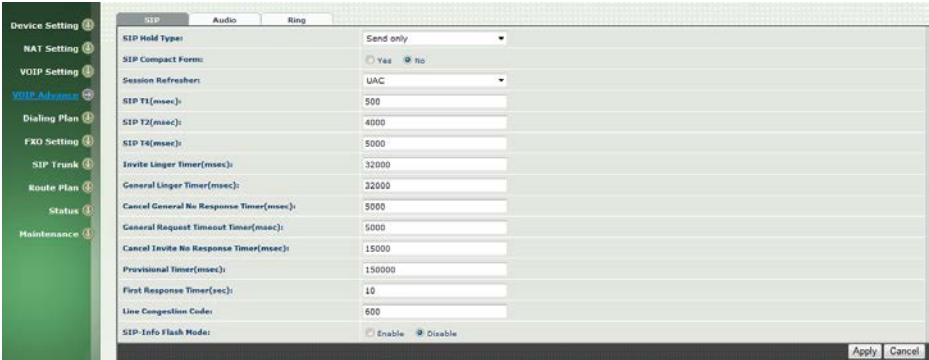


Figure 5-1 SIP

Parameter Description:

- **SIP Hold Type:** SIP on hold message sending method.
 - **Send Only:** Set the SDP media to sendonly when send an on-hold SIP message.
 - **0.0.0.0:** Set the SDP connection to 0.0.0.0 when send an on-hold SIP message.
 - **Inactive:** Set the SDP media to inactive when send an on-hold SIP message.
- **SIP Compact Form:** Enable SIP compact form or not. When enable this feature, the connected SIP proxy is required to support compact form.
- **Session Refresher:** Who will send dialog keep alive message (re-invite or update).
 - UAC: User Agent Client will do the refresh (default setting)
 - UAS: User Agent Server will do the refresh
- **SIP T1 (msec):** T1 determines several timers as defined in RFC3261. For example, when an unreliable transport protocol is used, a Client Invite transaction retransmits requests at an interval that start at T1 seconds and doubles after every retransmission. A Client General transaction retransmits requests at an interval that starts at T1 and doubles until it reaches T2. (Default Value: 500ms)

* *

- **SIP T2 (msec):** Determines the maximum retransmission interval as defined in RFC3261. For example, when an unreliable transport protocol is used, general requests are retransmitted at an interval which starts at T1 and doubles until reaches T2. If a provisional response is received, retransmission continue but at an interval of T2. (Default Value: 4000ms) **
- **SIP T4 (msec):** T4 represents the amount of time the network takes to clear message between client and server transactions as defined in RFC3261. For example, when working with an unreliable transport protocol, T4 determines the time that UAS waits after receiving an ACK message and before terminating the transaction. (Default Value: 5000) **
- **Invite Linger Timer:** After sending an ACK for an INVITE final response, a client cannot be sure that the server has received the ACK message. The client should be able to retransmit the ACK upon receiving retransmissions of the final response for this timer. This timer is also used when a 2xx response is sent for an incoming Invite. In this case, the ACK is not part of the Invite transaction.
- **General Linger Timer:** After a UAS sends a final response, the UAS cannot be sure that the client has received the response message. The UAS should be able to retransmit the response upon receiving retransmissions of the request based on this timer.
- **Cancel General No Response Time (msec):** When sending a CANCEL request on a General transaction, the User Agent waits cancel General No Response Timer milliseconds before timeout termination if there is no response for the cancelled transaction(Default Value: 10000ms).**
- **General Request Timeout Timer (msec):** After sending a General request, the User Agent waits for a final response general Request Timeout Timer milliseconds before timeout termination (in this time the User Agent retransmits the request every T1, 2*T1,...T2,...milliseconds)**
- **Cancel Invite No Response Timer (msec):** When sending a CANCEL request on an Invite request, the User Agent waits this timer before timeout termination if there is no response for the cancelled transaction.
- **Provisional Timer (msec):** The provisionalTimer is set when receiving a provisional response on an INVITE transaction. The transaction will stop retransmissions of the INVITE request and will wait for a final response until the provisionTimer expires. If you set the provisionTimer to 0, no timer is set. The INVITE transaction will wait indefinitely for the final response.
- **First Response Timer (msec):** When sending a request out, the User Agent waits this timer for any response received from UAS. If timer is expired and no any SIP message is received, the User Agent

will think the request is failed. The default is 5 seconds.

- **Line Congestion Code:** when callee's end system was contacted successfully but the callee is busy and does not wish to take the call at this time, the system wills response the code, default is 600. (FXO use)
- **SIP-Info Flash Mode:** when you enable the feature, system will make flash key to send SIP message by sip-info.

5-2 Audio

The setting page includes the device related audio settings.



Figure 5-2 Audio setting

- **RFC 2833 Payload Type:** 96 or 101. It is recommended to use 101.
- **DTMF Send On Time(msec):** When generate DTMF, the DTMF on time will be send (default value is 70 ms)
- **DTMF Send Off Time(msec):** When generate DTMF, the DTMF off time will be send (default value is 70 ms)
- **DTMF Detect Min on Time (msec):** The minimum DTMF on time will be processed as a regular DTMF event. Smaller than it will be ignored. The default value is 60ms.
- **DTMF Detect Min off Time (msec):** The minimum DTMF off time for the same DTMF value. Smaller than it and the new DTMF digit is the same as previous one will be handled as 1 digit only.
- **DTMF Relay Volume:** The DTMF relay volume
- **T.38 Fax Volume:** The T.38 fax relay volume
- **T.38 Redundant Depth:** The T.38 redundant packet depth. It could 0 (no redundant), 1 or 2. It is recommended to set to 2.
- **T.38 ECM:** The t.38 error correction mode. Default value is ON.

- **Min Jitter Buffer (msec):** The minimum delay time of Jitter buffer.
- **Max Jitter Buffer (msec):** The Maximum delay time of Jitter buffer.
- **Max Echo Tail Length (G.168):** Enable the echo cancellation feature. The default setting is "128ms".
- **Jitter Opt. Factor:** Jitter buffer dynamic factor for optimize. Please set to 7 unless under Welltech's instruction to change.
- **Impedance:** selected analog phone's impedance. (for FXS port use)

5-3 Ring

When you connect WellGate 2540 FXO gateway Lines to local PABX's extension or PSTN switch, please configure proper [Ring Cadence \(Ring ON time and Ring OFF time\)](#) according to [the PABX or PSTN ring Cadence specification](#).

Without configuring Ring Cadence to FXO gateway properly according to PABX or PSTN Ring Cadence, FXO may detect Ring signal in a strange behavior and cause abnormal operation.

WellGate 2540 factory default Ring Cadence setting is 1 seconds ON, 2 seconds OFF. Some PABX or PSTN provides 1 second ON and 3 seconds OFF Ring Cadence. In this case, please configure WellGate 2540 Ring Cadence to 1 second ON and 3 seconds OFF to match incoming Ring signal. (See Figure 5-3 Ring setting.)

Note: if you have no idea what is the PABX or PSTN ring cadence, please to change the ring on and ring off to **8000**. When have an incoming call which WellGate 2540 can't pick up it.



Figure 5-3 Ring setting

- ◆ **Frequency (10~70HZ):** Specify the ringing frequency value

(default is 20HZ)

- ◆ **Ring on (0~8000ms)**: Specify the ringing on value (default is 1000msec)
- ◆ **Ring off (0~8000ms)**: Specify the ringing off value (default is 2000msec)
- ◆ **Ring level (10~95volt)**: Specify the ringing level (default is 94 volt)

CH6 Dialing Plan

6-1 General



Figure 6-1 General setting

- **First Digit Time Out:** Specify the duration of dial waiting when the receiver is off hook. The range is 1~60 sec.
- **Inter Digit Time Out:** Specify the interval of input digits, if the interval is over the setting, the system will end the dial and send out the DTMF. The limitation range is 1~10sec.
- **End of Digit:** The assigned key will be tread as end of dial.
- **Retrieve Number:** it will forced to get back line, if used Welltech 2540 make transfer to other devices but the devices no answer and into voice mail, You can press the code forced to get back line. Default is "***#**".

6-2 Dialing Rule



Figure 6-2 Dialing Rule setting

Dialing rule is used to speed up the dialing procedure. Some user don't like to use the end of dialing digit such as "#", the administrator can use dialing rule instead. The longest prefix will be matched first.

- **Dialed Prefix:** The prefix to be matched
- **Max Digits:** The digits will be received based on the Dialed Prefix.

The following is an example for dialing rule:

Mobile call is started with 09 and it is 10 digits

Long distance call is started with 0 and it is 10 digits

International call is started with 00 and its max digit should be less than 32

The others are local call and 8 digits

Emergency call is started with 1 and 3 digits

The Dialing rule can be set as follows:

Prefix, max digits

09, 10

0, 10

00, 15

1, 3

2, 8

3, 8

4, 8

5, 8

6, 8

7, 8

8, 8

9, 8

6-3 Digit Manipulation

The Digit Manipulation will be processed based on prefix and DM group after the DNIS is determined.



Figure 6-3 Digit Manipulation setting

- **DM Group:** Different DM group have different case to be used.
 - ◆ **FXO:** This DM group is used for FXO 2 stage dialing. After the DNIS is collected, this DM group will be processed before enter the routing procedure.
 - ◆ **VOIP:** This DM group is used for VOIP incoming call. After the DNIS is collected in 2 stage dialing or 1 stage dialing DNIS, this DM group will be processed before enter the routing procedure.
 - ◆ **1-4:** These DM groups are used for backup routing purpose. When a backup routing is used, the administrator can select a DM group to be processed before start the backup route.
- **Matched Prefix:** The prefix to be matched for DM. The longest prefix will be matched first.
- **Matched Length:** Set to 0 for ignoring the length. The other 1-32 are the length to be matched as a condition.
- **Start Pos:** The start position to be replaced.
- **Stop Pos:** The stop position to be replaced.
- **Replace Value:** The value to replace.

Example of Digit Manipulation Settings:

Prefix	Len	Start Pos	Stop Pos	Replace Value	Test DNIS	Result DNIS
886	0	0	0	002	8862123456	0028862123456
886	12	0	0	002	8862123456	8862123456
886	0	2	5	002	8862123456	8800223456
886	0	30	30	002	8862123456	8862123456002
886	0	1	6		8862123456	83456

6-4 Phone Book

Phone Book is used for peer to peer call.



Figure 6-4 Phone Book setting

- **Name:** This field supports **called number** only. If you enter words or text here, it will routes to proxy server automatically.
- **Tel No:** Enter called number and IP address. Please follow this sample of picture, as the format of "number@uri:port". (default port is 5060)
- **Export:** To backup the phone book records.
- **Import:** To reload setting of phone book.

CH7 FXO Setting

The FXO Setting contains the FXO related parameters.



The screenshot shows a web-based configuration interface. On the left is a vertical navigation menu with various settings categories: Device Setting, NAT Setting, VOIP Setting, VOIP Advance, Dialing Plan, FXO Setting (highlighted in blue), SIP Trunk, Route Plan, Status, and Maintenance. The main content area is titled 'FXO line' and contains a table with the following data:

	Line ID	State	TEL No	Hot Line TEL
	1	Active		
	2	Active		
	3	Active		
	4	Active		

Figure 7-0 FXO setting

- **Line ID:** FXO line (L1 to L4)
- **State:** The line is active or not
- **TEL No:** The reference telephone number (e.g. PSTN TEL of line)
- **Hotline TEL:** If hot line is set, this field shows the hot line number.
- **Export:** backup all lines setting.
- **Import:** reload all line setting.

7-1 FXO line

Modify Line Setting	
Line ID :	1
Line Type :	FXO
Line State :	<input checked="" type="radio"/> Active <input type="radio"/> Inactive
TEL No :	
Polarity Reversal Detection :	<input type="radio"/> Yes <input checked="" type="radio"/> No
Current Drop for Disconnect :	<input type="radio"/> Yes <input checked="" type="radio"/> No
Incoming Call Handling :	2 Stage Dialing
Hot Line TEL :	
Playback Voice File :	<input type="radio"/> Yes <input checked="" type="radio"/> No
Repeat Count :	
Voice File Name(HuLaw:mono 0K):	
Flash Time(msec):	300
FAX Relay :	T.38
Input[Encode] Gain:	0dB
Output[Decode] Gain:	0dB
Dialing Answer Delay Time(sec):	3
PSTN Answer Ring Count:	2
Caller ID Mode :	None

Figure 7-1 FXO setting

- **User ID:** FXO Line number (L1 to L4)
- **User Type:** the line type which is FXO
- **Line State:** Set to active if you would like to use this line. Otherwise, set to Inactive.
- **TEL NO:** This field can be used as a reference remark for this line. Normally, you can put the connected PSTN line's phone number here for reference.
- **Polarity Reversal Detection:** When enable the Polarity Reversal Detection; the Welltech 2540 will use the polarity reversal signal as the answer signal for FXO outgoing call. When disable the polarity Reversal Detection, the Welltech 2540 will use Dialing Answer Delay Time for answering the SIP call.
- **Current Drop for disconnect:** Use Line current drop as a disconnect supervision or not.
- **Incoming call handling:** The call handling policy for a FXO incoming call.
 - ◆ **Hot line TEL:** When a FXO incoming call was detected and after the PSTN Answer Ring Count, Welltech 2540 will send the SIP call to the specified hot line TEL number through the Route Plan.
 - ◆ **2 Stage Dialing:** When a FXO incoming call was detected and after the PSTN Answer Ring Count, Welltech 2540 will answer it and play the dial tone for 2 stage dialing to VOIP.
- **Playback voice file:** To enable playing voice greeting file or not.
- **Repeat Count:** Repeat how many counts to play voice greeting file.

(Used for FXO port with 2-Stage Dialing Only)

- **Voice file name (MuLaw-mono 8K):** Specify the file path and file name to upload. Please make sure that the file format needs to be **G.711U, 8K, 8 bits raw file. (Used for FXO port Only)**
- **Flash Time:** Flash Time will be send to PSTN line (internal use only)
- **FAX Relay:** Enable T.38 Fax Relay or not
- **Input(Encode)Gain:** Adjust the volume from PSTN to VOIP (default is 0 db)
- **Output(Decode)Gain:** Adjust the volume from VOIP to PSTN (default is 0 db)
- **Dialing Answer Delay Time (sec):** When the polarity reversal detection is disabled, Welltech 2540 will answer the call after this time out without answer supervision. After the DTMF dialing, Welltech 2540 will send 183 with SDP to enable the voice path for VOIP side.
- **PSTN Answer Ring Count:** This ring count is used for called ID detection and 2 stage dialing.
 - If the caller ID is sending during the first ring and second ring, this parameter should be set to greater or equal to
 - If the caller ID is sending before the first ring, this parameter can be set to greater or equal to 1.After the ring count was reached, Welltech 2540 will answer the call if 2 stage dialing is selected or make the VOIP call out if hot line is selected.
- **CallerID Standard:** The detected Caller ID stand for the line.

CH8 SIP Trunk

The administrator needs to set the SIP trunk for VOIP outgoing call and incoming call. There are up to 4 SIP trunk can be used for whole system.



Trunk ID	Register Type	TEL No	Proxy Server	Proxy Server Port	Outbound Proxy	Outbound Server Port
1	Register	1001	192.168.18.247	5060		5060

Figure 8-0 SIP Trunk page

- **Trunk ID:** SIP trunk ID 1 to 4
- **Register Type:** Register type is predefined or register
- **TEL No:** The Tel no for the SIP account
- **Proxy Server:** The SIP proxy server
- **Proxy Server port:** The SIP proxy server port
- **Outbound Proxy:** The SIP outbound proxy sever
- **Outbound Server Port:** The SIP outbound proxy server port

8-1 Create SIP Trunk

The screenshot shows the 'Modify SIP Trunk' configuration page. The left sidebar contains navigation options: Device Setting, NAT Setting, VOIP Setting, VOIP Advance, Dialing Plan, FXO Setting, SIP Trunk (highlighted), Route Plan, Status, and Maintenance. The main configuration area includes the following fields:

- Trunk ID: 1
- Register Type: Register Predefine
- Domain: [Empty]
- Proxy Server: 192.168.18.247
- Proxy Server Port: 5060
- Outbound Proxy Server: [Empty]
- Outbound Proxy Server Port: 5060
- Register Expires: 120
- TEL No: 1001
- User ID: 1001
- User Password: ****
- Display Name: 1001
- Reject Anonymous Call: Yes No
- Outgoing Caller ID:
 - Display Name: PSTN Caller ID
 - User ID: PSTN Caller ID
- For DRIS is Register TEL: 1 Stage Dialing 2 Stage Dialing
- Keep Alive: Enable Disable
- Keep Alive Time (sec): [Empty]

Buttons for 'Apply', 'Cancel', and 'Back' are located at the bottom right of the form.

Figure 8-1 SIP Trunk page

- **Trunk ID:** SIP trunk ID 1-4
- **Register Type:** Whether this account need register or not
 - ◆ **Register:** When it is set to register, Welltech 2540 will send REGISTER message to SIP proxy server for registration.
 - ◆ **Predefine:** When it is set to predefine, Welltech 2540 will NOT send REGISTER message out.
- **Domain:** The SIP domain for register or call making
- **Proxy Server:** SIP registrar server address
- **Proxy Server Port:** SIP registrar server port number
- **Outbound Proxy Server:** outbound proxy server address
- **Outbound Proxy server port:** outbound proxy server port number
- **Register Expires:** the default register expires for negotiation
- **TEL No:** The registrar telephone number
- **User ID:** The SIP user ID for register and call making
- **User Password:** The SIP password for register and call making
- **Display Name:** The SIP display name
- **Reject Anonymous Call:** Reject the anonymous call
- **Outgoing Caller ID:** The outgoing SIP caller ID mode.
 - Display Name:** The display name will be set according to the following type.
 - None:** No display name will be used
 - PSTN caller ID:** The display name will be the collected PSTN caller ID

SIP display name: The display name will be the Display Name set in this SIP trunk.

FXO Tel NO: The display name will be the incoming FXO's TEL No set on FXO lines.

User ID: The SIP caller ID will be used according to the following type.

- **SIP user ID:** If the SIP user ID is set, the SIP user ID set in this SIP trunk will be used and the domain/SIP proxy will be the host part. The SIP FROM header's URL will be the [SIP_User_ID@Domain](#) or [SIP_User_ID@SIP_Proxy_Server](#).
- **PSTN caller ID:** If the PSTN caller ID will be used in SIP URL, the SIP FROM header's URL will be [PSTN_Caller_ID@local_IP_address](#).
- **FXO Tel NO:** If the FXO Tel NO will be used in SIP URL, the SIP FROM header's URL will be [FXO_Tel_NO@local_IP_address](#).

The following guideline could be used for most of case:

1. If the Welltech 2540 in SIP proxy is handled as a gateway, please set the display name and User Id should be set to "PSTN caller ID".
2. If the Welltech 2540 is SIP proxy is handled as a subscriber, please set to the display name to "PSTN caller ID" and user ID to "SIP User ID".

- **For DNIS is Register TEL:** When you have a call from VoIP to FXO to call out to PSTN network, there are two methods can be used.
- **Keep Alive:** Enable or Disable it.
- **Keep Alive Time (sec):** Specify of times send sip register message to proxy server.

CH9 Route Plan

The core of Welltech 2540 is the routing policy. The policy is based on incoming call type/target, length and prefix to determinate the outgoing call process. For VOIP incoming call, it can send to FXO interface and vice versa.



Incoming Call Type	Matched Prefix	Matched Incoming List	Matched Length	Outgoing Type
VOIP Default Route		1,2,3,4	0	FXO
FXO Default Route		Line 1,2,3,4	0	VOIP

Figure 9-0 Route Plan page

- **Incoming Call Type:** Incoming call type (FXO or VOIP)
- **Matched Prefix:** matched DNIS (called number) prefix
- **Matched Incoming List:** matched DNIS incoming interface target
- **Matched Length:** matched DNIS (called number) length
- **Outgoing Type:** The outgoing call type (FXO or VOIP)
- **Export:** backup route plan setting.
- **Import:** reload route plan setting.

9-1 Create Route Plan

Click Route Plan and Click new to create a new routing policy.

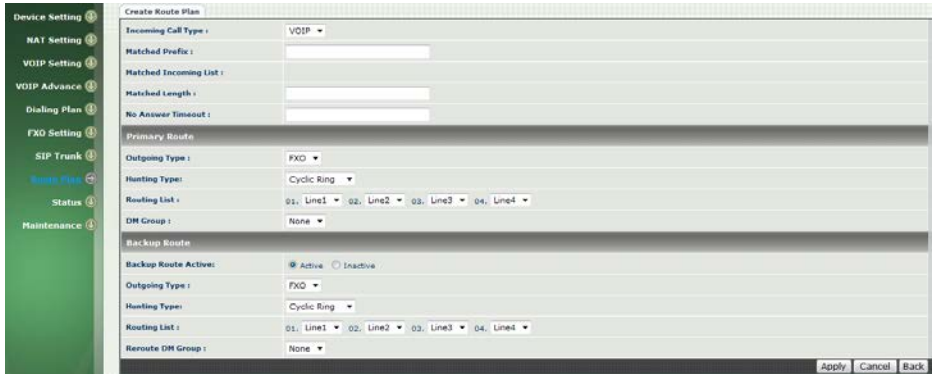


Figure 9-1 Route Plan setting

- **Incoming Call Type:** Incoming call type
 - **VOIP:** The incoming SIP call type
 - **FXO:** The PSTN incoming call type
- **Matched Prefix:** matched DNIS (called number) prefix
- **Matched Incoming List:** matched DNIS incoming interface target
 - For VOIP incoming call type, the incoming target will be the SIP trunk ID. Only the call from the selected SIP Trunk will be accepted for this route.
 - For PSTN incoming call type, the incoming target will be the line ID (L1 to L24). Only the call is coming from the selected line will be accepted for this route.
- **Matched Length:** matched DNIS (called number) length. For ignoring the length, please set to 0.
- **No Answer Timeout:** How long the hunting will continue to next when the called target doesn't answer.

Create Route Plan>Primary Route

- **Outgoing Type:** Outgoing call type (FXO or VOIP)
- **Hunting Type:** The hunting method will be used for this route.
 - **Priority Ring:** The call will be hunted based on the routing list order one by one.
 - **Cyclic Ring:** The call will be hunted based on the cyclic

basis. This is the recommended method.

- **Routing List:**
 - The routing target list will be used for this route.
- **DM Group:** Select DM group 1 to 4 in case it requires an DM (for example remove the prefix) before to make the call.

Create Route Plan>Backup Route

- **Backup Route Active:** Active the backup route or not.
- **Outgoing Type:** The backup route outgoing call type.
- **Hunting Type:** The hunting method will be used for this route. Please refer to the Primary Route.
- **Routing List:** The backup routing target list will be used for this route.
- **Route DM Group:** Select DM group 1 to 4 in case the backup required the DM before to make the call. The DNIS is unchanged by the primary route DM and same as the DNIS before routing. For example, the DNIS is 886282265699 and primary DM group remove 886 and use it (DNIS = 282265699) to make call. When backup route is started, the DNIS is still unchanged as 886282265699. This makes the DM easy to predict and implement.

2 special default route, "VOIP Default Route" and "FXO default Route", are used as the default routing when there is no any other routing are matched. It is not recommended to disable these 2 default route. The FXO default route is used when a FXO incoming call's default routing. VOIP default route is used for a VOIP incoming call's default routing.

CH10 Status

Welltech 2540 provides the system status here.



The screenshot shows the 'Device Status' tab in the Welltech 2540 web interface. The left sidebar contains navigation options: Device Setting, NAT Setting, VOIP Setting, VOIP Advance, Dialing Plan, FXS Setting, SIP Trunk, Route Plan, and Maintenance. The main content area displays the following information:

Device Status	Line Status	SIP Trunk Status
Model:	WQ2540	
MAC-Address:	WAN: 00:01:A8:08:5C:E8 LAN: 00:01:A8:08:5C:E7	
Network Type:	DHCP	
IP-Address:	WAN: 192.168.23.21 LAN: 192.168.123.123	
IPV6 IP-Address:		
Firmware:	hg25.2.03f.bin	

Figure 10-0 Device Status

10-1 Device States

[See the figure 10-0 Device Status](#)

- **Model:** The model number
- **MAC-Address:** The MAC address of Welltech 2540
- **Network Type:** The Network Interface Type Settings
- **IP-Address:** IP address is using
- **IPV6 IP-address:** display IPV6 address
- **Firmware:** The firmware version and release information

10-2 Line States

This page shows the each line's current status.



Line	Call State
1	Not Connected
2	Not Connected
3	Not Connected
4	Not Connected

Refresh Interval (second): 5

Figure 10-2 Line Status

- **Line:** L1 to L4
- **Call State:** The Line status for this line
- **Refresh Interval (second):** The time to refresh the status

10-3 SIP Trunk States



Account	Registered	Concurrent Call
0702062424	Not Register	0
0700343678	Registered	0
0702060403	Registered	0

Refresh Interval (second): 1

Figure 10-3 SIP Trunk Status

- **Account:** SIP trunk account
- **Registered:** The SIP trunk register status
- **Concurrent Call:** The concurrent calls are used for this SIP trunk
- **Refresh Interval (second):** The time to refresh the status

CH11 Maintenance

Welltech 2540 can be managed by this management page for upgrading firmware or reset.



Figure 11-0 Maintenance

- **Backup:** Backup the system settings for restoring purpose
- **Restore:** Restoring the backup setting back to Welltech 2540
- **Reset to Default:** Reset system setting to factory default
- **Quick-Reset:** Warm Reset without reboot Welltech 2540
- **Reboot:** reboot Welltech 2540

11-1 Firmware Update

This maintenance page provides the firmware upgrade features.



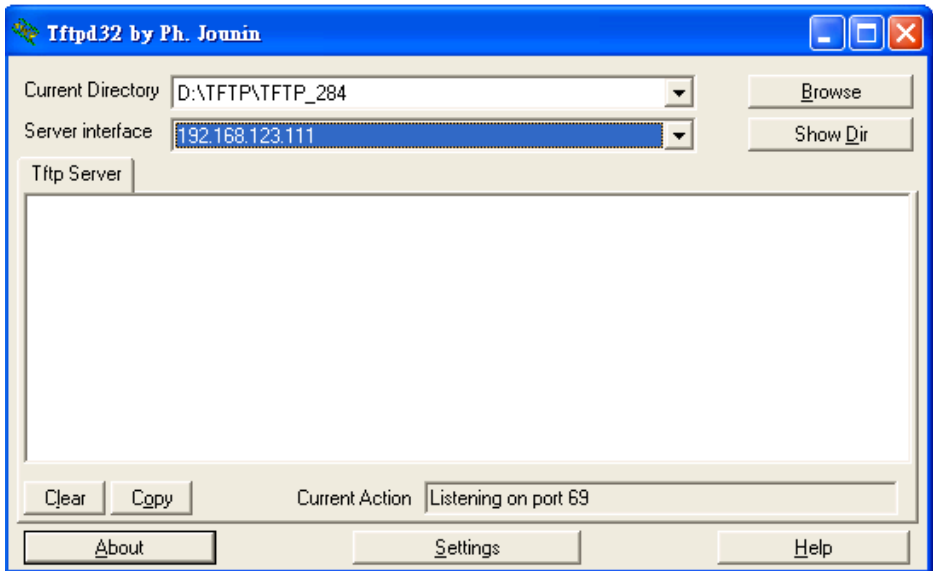
Figure 11-1 firmware update

- **Firmware Update:** Upgrade the new firmware through web page

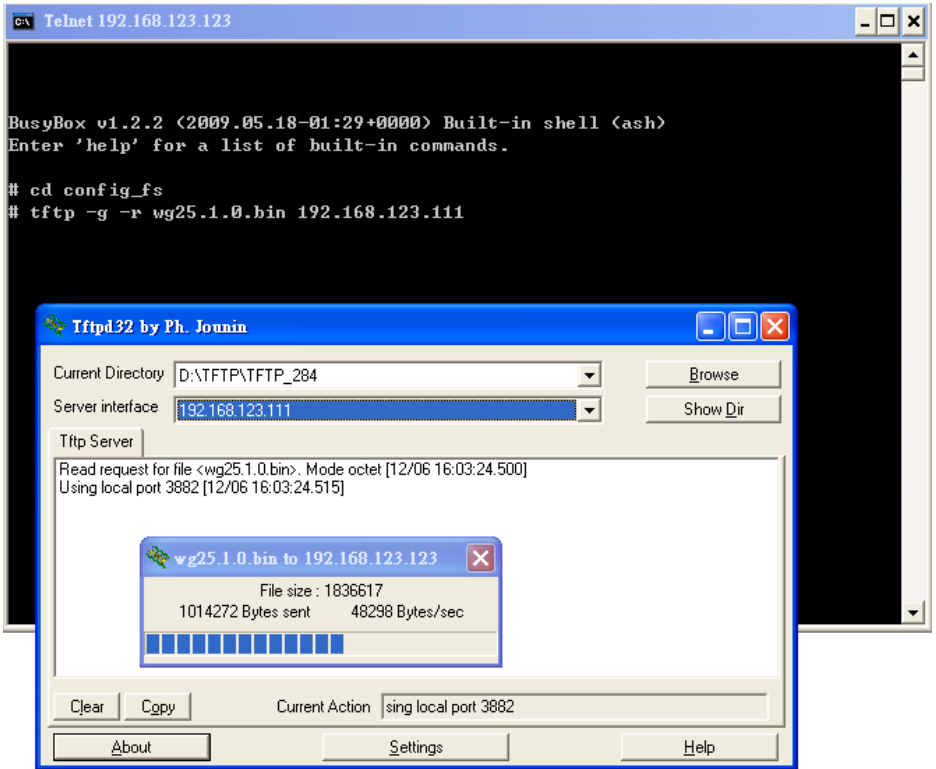
Appendix A --- System Recovery

WG2540 use dual firmware image to ensure the system stabilities. In most of case, you will not encounter the system failed to boot issue. Normally, the user should be able to use Web page to login and upgrade the firmware through it. If you are not able to do it, please follow the following steps for recovery.

1. Start the WG2540 and to check the STATUS led is up or not. If STATUS led is ON, please press the reset button for 5 seconds to reset to default. After all LED are light up, the system is back to factory settings.
2. Change your PC IP address to **192.168.123.111** and network set to fix IP address mode.
3. Connect your PC to WAN port and use <http://192.168.123.123> to upgrade the firmware. Make sure you are using Microsoft IE 6 or later version. **Do not support FireFox Web browser.**
4. If you cannot login to the web page through **192.168.123.123**. Open a command line windows and type "telnet 192.168.123.123". If you can see the following display, go to the next step. Otherwise, please contact Welltech FAE for RMA (Repair).
5. Prepare a TFTP server for firmware download as follows
 - download tftp server
http://www.welltech.com/support/voip/TFTP/TFTP_Server.zip
or
http://tftpd32.jounin.net/tftpd32_download.html
 - start tftp server



- download the firmware into tftp data directory
6. In the telnet terminal, do the following command
- 1. __dmctw
 - 2. cd /config_fs
 - 3. rm -f wg25*.bin
 - 4. tftp -g -r wg25.1.0.bin 192.168.123.111



- 4. copy firmware successfully
 - 5. reboot
7. Check whether the system was recovered or not

Appendix B --- HTTP auto provisioning

Get the http provisioning packet from Welltech and start the provisioning as follows:

Step 1: build mac list for mass configuration file generation

For FXO>

	A	B	C	D	E	F	G	H	I	J	K	L	M	N
1	\$MACAddress	sipdk1.displayname	sipdk1.userid	sipdk1.password	sipdk1.telno	sipdk2.displayname	sipdk2.userid	sipdk2.pass	sipdk2.telno	sipdk3.displayname	sipdk3.userid	sipdk3.pass	sipdk3.telno	sipdk4.displayname
2	0001a800de35	1001	1001	1001	1001	1002	1002	1002	1002	1003	1003	1003	1003	1
3														
4														

The Welltech 2540 MAC.csv contains most frequently changed parameters as following:

MACAddress: WellGate 2540 MAC Address

Sipdk1.displayname ~ sipdk4.displayname: display name for each line

Sipdk1.userid ~ sipdk4.userid: user id for register to SIP proxy for each line

sipdk1.password ~ sipdk4.password: user password for register to SIP proxy for each line

Sipdk1.telno ~ sipdk4.telno: tel no for each line

Please save and close it.

Step 2: create a template configuration file

Open the "Welltech 2540 Parameter.txt" getting from Welltech and make the required change. Please at least make the changes for those provisioning and SIP proxy settings. [For detail, please refer the comments of "wg2540 Parameter.txt"](#).

Step 3: Make the change for wegencfg.ini as follows if necessary

```
# Template File
BaseFile=.\Welltech 2540 Parameter.txt
# MAC list file
ListFile=.\Welltech 2540 MAC.csv
# 0: Off, 1: On
Encrypt=0
```

Step 4: Generate the individual configuration file.

Double click the "wtgencfg.exe", it will generate the configuration file for each MAC list in "MAC address.cfg" as the following pictures.

```
wtgencfg.exe 1.0.0 (R091006)
18:02:09 L0114 Notice : wtgencfg.exe 1.0.0 <R091006> start.
18:02:09 L0270 Info : Read "\.\\wtgencfg.ini".
18:02:09 L0271 Info : [System].
18:02:09 L0289 Info : Action=0.
18:02:09 L0304 Info : ToCase=0.
18:02:09 L0315 Info : BaseFile=\\.\\wg2540 Parameter.txt.
18:02:09 L0327 Info : ListFile=\\.\\wg2540 MAC.csv.
18:02:09 L0340 Info : HeadTagPrefix=FmtNo=_wtcfg_
18:02:09 L0348 Info : HeadTagSuffix=wg25xx.
18:02:09 L0355 Info : Encrypt=0.
18:02:09 L0364 Info : Default EncryptKey=*****.
18:02:09 L0397 Info : Read "\.\\wg2540 Parameter.txt" 30772 element(s).
18:02:09 L0454 Info : Read "\.\\wg2540 MAC.csv" 355 element(s).
18:02:09 L0665 Info : Opened "\.\\0001a800de35.cfg".
18:02:09 L0703 Info : "\.\\0001a800de35.cfg" finished 290 record(s).
18:02:09 L0563 Info :
18:02:09 L0154 Notice : Total 1 file(s) generated.
請按任意鍵繼續 - -
```

Step 5:

Put the "*.cfg" file into http or ftp directory. Set the provisioning settings in WellGate 2540 and reboot to test it. You can use the hfs for http file server. It can be download from <http://www.rejetto.com/hfs/>.

Note: please link it to download provision file. More information please refers "wg2540 Parameter.txt".

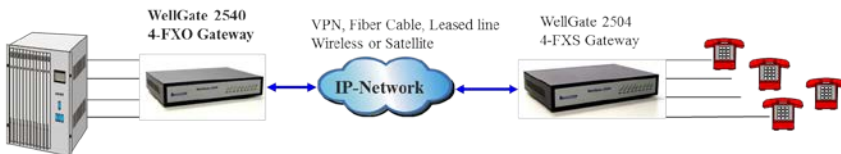
http://www.welltech.com/support/voip2/SIP%20series/FXSO%20series/25xx/provision/2540/2540_Provision.zip

Appendix C--- WellGate 2504 (4-FXS) and WellGate 2540(4-FXO)

in peer to peer mode with hotline by port to port application.

Purpose

This paper is going to describe how to configure FXO to FXS in **peer to peer mode with hotline function** to link from one desire port at FXO via SIP IP link to remote FXS desire port. This application is widely used to extend traditional PABX extension from one location to remote office via IP Link which could be VPN, Private network, fiber link or Wireless network. Here is an typical diagram of this application at Figure 1.



	PABX Extension Number or PSFN number	FXO Telephone Number (default is blank)	FXO SIP Trunk Line Number	FXS Line Number	FXS Phone Caller ID number when FXO Select SIP Display number	FXS Phone Caller ID number when FXO Enable Caller ID Detection
Line 1	201		1001	501	1001	201
Line 2	202		1002	502	1002	202
Line 3	203		1003	503	1003	203
Line 4	204		1004	504	1004	204

Figure 1.

Before start to configure of both FXO and FXS gateway, we have assigned the following parameters to both devices as follows. Please note that both IP addresses at FXO and FXS gateway's WAN port should be able to see each other. Both FXO and FXS gateway **CAN NOT** install behind Router or Firewall or IP Sharing.

WellGate 2504 (4-FXS Gateway) information :

IP address : 192.168.18.201

Line number : 501, 502, 503 and 504

WellGate 2540 (4-FXO Gateway) information :

IP address : 192.168.23.23

SIP Line number : 1001, 1002, 1003 and 1004

FXO Telephone Number : default is Blank

PABX Extension number 201, 202, 203 and 204 which were connected to Port #1 to #4 at FXO Gateway.

WellGate 2504 (FXS Gateway) configuration

Step 1: Configure static IP address to WellGate 2504, like below [Figure 2](#). For example: Set static IP address of 192.168.18.201 at **WAN port** on **Device Setting** → **Network** of WellGate 2504 WEB page.



Figure 2.

Step 2: Go to **VOIP Setting** → **SIP** to select the “**Accept Proxy Only**” feature to **NO** and make sure to setup the **SIP Local Port** number. In this example, the SIP local port is 8080. See [Figure 3](#). This port number should be the same as Phone book configuration SIP port number.



Figure 3.

Step 3: Set the FXS line's information to each port (Line ID is the FXS's port number at [Figure 4](#)).

Enable Hotline feature at FXS gateway to port #1 and enter **HOT LINE TEL**

field to **1001** which is WellGate 2540 FXO port 1 number. This command let FXS port #1 point to remote FXO port #1 directly. Continue to configure FXS port number 2 to 4 as following example at **Modify Line Setting** web page. See [Figure 5](#).

Enter **HOT LINE TEL** field to **1002** which is WellGate 2540 FXO port 2 number. (This command let FXS port #2 point to remote FXO port #2 directly.)

Enter **HOT LINE TEL** field to **1003** which is WellGate 2540 FXO port 3 number. (This command let FXS port #3 point to remote FXO port #3 directly.)

Enter **HOT LINE TEL** field to **1004** which is WellGate 2540 FXO port 4 number. (This command let FXS port #4 point to remote FXO port #4 directly.)

Step 4: Go to **Modify Line Setting** webpage (See [Figure 5](#)) of FXS gateway at **Port #1 (LINE ID 1)** to continue setting of port #1.

Select **Register Type** to “**predefine**”. Enter FXS port #1 information as follows.

TEL No : 501

User ID : 501

User password: 501

Display Name: 501

Enter FXS port #2 information as follows.

TEL No : 502

User ID : 502

User password: 502

Display Name: 502

Enter FXS port #3 information as follows.

TEL No : 503

User ID : 503

User password: 503

Display Name: 503

Enter FXS port #4 information as follows.

TEL No : 504

User ID : 504

User password: 504

Display Name: 504

Step 5: Select **HOT Line** to **Enable** at webpage ([See Figure 5](#)) of Line Setting at FXS Gateway. And enter **HOT Line TEL** at **1001** which is WellGate 2540 FXO port #1 number.

For example:

The FXS port 1 number is 501, FXS port 2 is 502, FXS port 3 is 503, FXS port 4 is 504.

FXS port 1 number 501 hotline to FXO SIP Trunk 1 which number is 1001,

FXS port 2 number 502 hotline to FXO SIP Trunk 2 which number is 1002,
 FXS port 3 number 503 hotline to FXO SIP Trunk 3 which number is 1003,
 FXS port 4 number 504 hotline to FXO SIP Trunk 4 which number is 1004,
See Figure 4.

FXS Line	SIP Proxy	Caller ID	Others
		Line ID	State
		TEL No	Hot Line TEL
		1	Active
		2	Active
		3	Active
		4	Active

Figure 4.

Modify Line Setting

Line ID : 1

Line Type : FXS

Line State : Active Inactive

Forward Reason : Unconditional Busy No Answer

Forward TEL :

No Answer Timeout(sec): 120

Call Waiting : Enable

Reject Anonymous Call : Yes No

Hot Line: Enable Disable

Hot Line TEL : 1001

Polarity Reversal Generation : Yes No

Current Drop Generation : Yes No

Input(Encode) Gain: 0db

Output(Decode) Gain: 0db

FAX Relay : T.38

Voice Mail Subscription: Enable Disable

Caller ID Mode : Transparent

SIP Caller ID Mode : Transparent

Register Type : Predefine

TEL No: 501

User ID: 501

User Password: ***

Display Name: 501

Apply Cancel Back

Figure 5.

Step 6: set the SIP Proxy information on **FXS Setting** → **SIP Proxy**. See Figure 6.

For example: Set the SIP Proxy Server's IP address at 192.168.23.23, Proxy port number at 8080 and Outbound Proxy server port number at 8080 as well which were FXO's IP address and SIP port number.

[Note: You have to fill in remote FXO Gateway's IP address and port number at this step. See figure 6.](#)

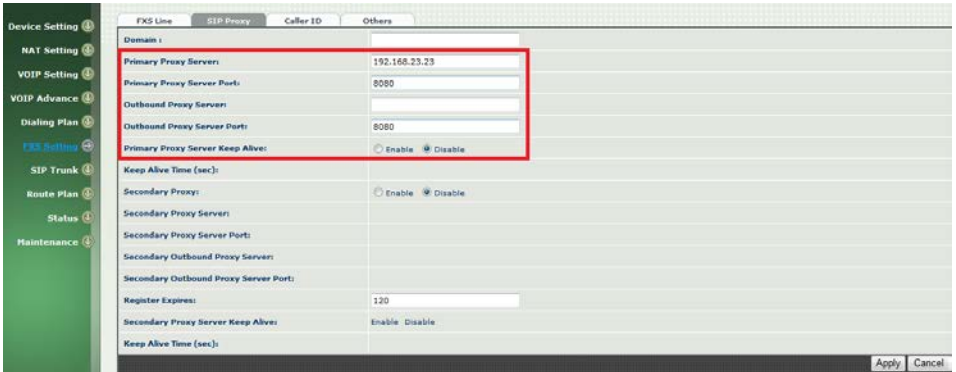


Figure 6.

Step 7: set WellGate 2540 FXO number and IP address on FXS gateway's webpage **Dialing Plan** → **Phone Book**. See Figure 7.

For example: the WellGate 2540 FXO port 1 number is 1001, SIP local port is 8080. The FXS gateway's phone Book format is as follows.
 FXO_line_number@FXO_ip_address:sip_port number.

At this example, the FXS gateway phone book should configure four ports as follows.

- Port 1 configure to [1001@192.168.23.23:8080](#)
- Port 2 configure to 1002@192.168.23.23:8080
- Port 3 configure to 1003@192.168.23.23:8080
- Port 4 configure to 1004@192.168.23.23:8080

Note:

The Name field of FXS Gateway's phone book should be remote FXO gateway's port number as 1001 to 1004. **Don't key in text format here.** It should be same as the remote FXO gateway's port number.



Figure 7.

WellGate 2540 (FXO Gateway) configuration

Step 1: set static IP address to WellGate 2540 FXO gateway's WAN port as following picture. See Figure 8. For example: Set static IP address 192.168.23.23 on **Device Setting** → **Network** of WellGate 2540 WEB page.

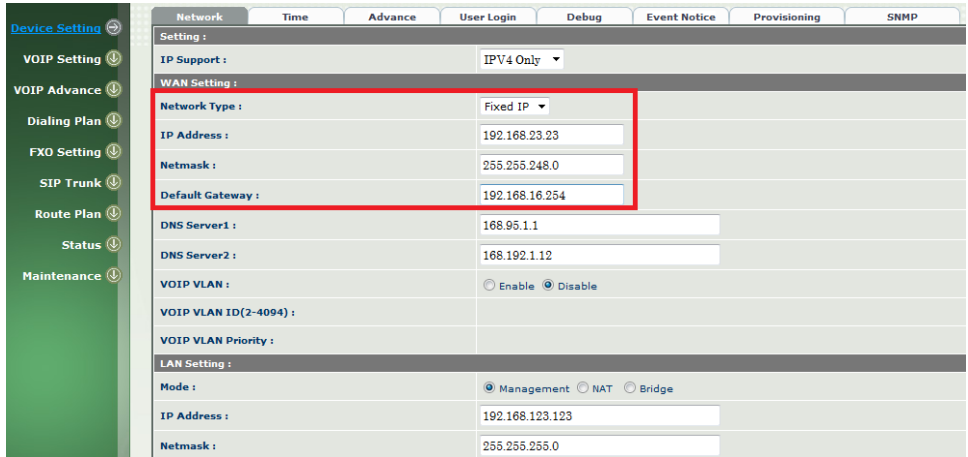


Figure 8.

Step 2: Go to **VOIP Setting** → **SIP** to select the “Accept Proxy Only” feature to **NO** and make sure to setup the **SIP Local Port** number to 8080. See Figure 9.



Figure 9.

Step 3: Set the FXO SIP line's information to each line. Go to **FXO Setting** webpage of WellGate 2540 FXO Gateway and configure each line by selecting Line ID from 1 to 4. See Figure 10. Enable **Hotline feature** by selecting **Incoming Call Handling** to **Hot Line** and give remote WellGate 2504 FXS gateway port 1 number to 501. For example: FXO SIP port 1 number 1001 hotline to remote FXS port 1 number 501 directly.

FXO SIP port 2 number 1002 hotline to remote FXS port 2 number 502 directly.
 FXO SIP port 3 number 1003 hotline to remote FXS port 3 number 503 directly.
 FXO SIP port 4 number 1004 hotline to remote FXS port 4 number 504 directly.

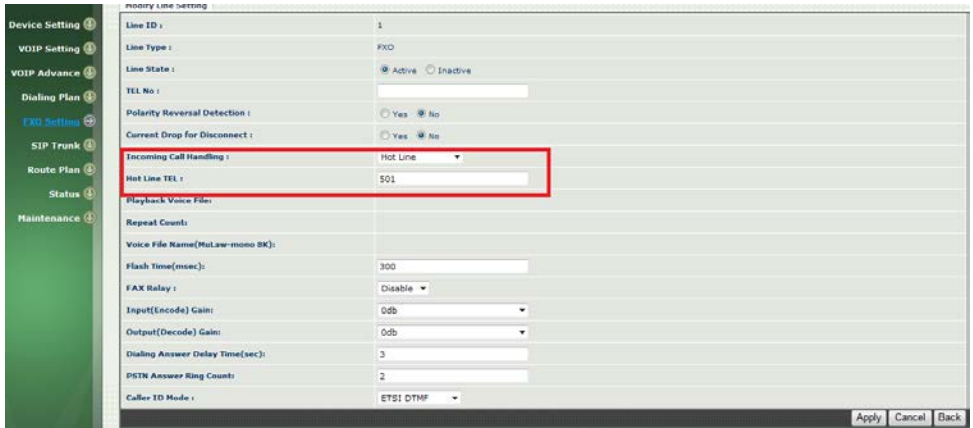


Figure 10.

Step 4: Set the **SIP Trunk** information to WellGate 2540 by going to **SIP Trunk → Modify SIP Trunk** webpage. See Figure 11.

1. Select **Register Type** to **Predefine** and key in the **Proxy Server IP address at 192.168.18.201 and port 8080 which are remote FXS Gateway's IP address and SIP port number.**
2. Next, configure each SIP Trunk number, user ID, password and display name to four SIP Trunks as following examples. **See Figure 11.**

Enter SIP Trunk #1 information as follows.

TEL No : 1001
 User ID : 1001
 User password: 1001
 Display Name: 1001

Enter SIP Trunk #2 information as follows.

TEL No : 1002
 User ID : 1002
 User password: 1002
 Display Name: 1002

Enter SIP Trunk #3 information as follows.

TEL No : 1003

User ID : 1003
User password: 1003
Display Name: 1003

Enter SIP Trunk #4 information as follows.

TEL No : 1004
User ID : 1004
User password: 1004
Display Name: 1004

3. Select **"2 Stage Dialing"** at **"DNIS is Register TEL"** webpage. When user picks up the handset from FXS port #1 of WellGate 2504, you should hear Dial Tone from PABX extension 201 which was connected to FXO Line #1. And it works like this analog phone was connected at remote PABX's extension 201. See Figure 11.

If set **"1 stage dialing"** in here, FXO will dial out line number 1001 to PABX automatically when user picks up the handset from FXS port #1 of WellGate 2504.

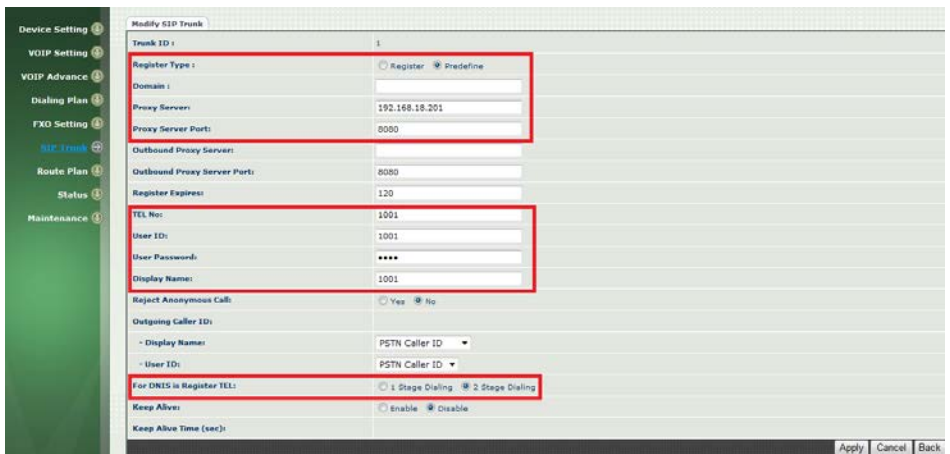


Figure 11.

Trunk ID	Register Type	TEL No	Proxy Server	Proxy Server Port	Outbound Proxy	Outbound Server Port
1	Predefine	1001	192.168.18.201	8080		8080
2	Predefine	1002	192.168.18.201	8080		8080
3	Predefine	1003	192.168.18.201	8080		8080
4	Predefine	1004	192.168.18.201	8080		8080

Step 5: Set Route Plan to each FXO line.
Disable the “VOIP Default Route” first.

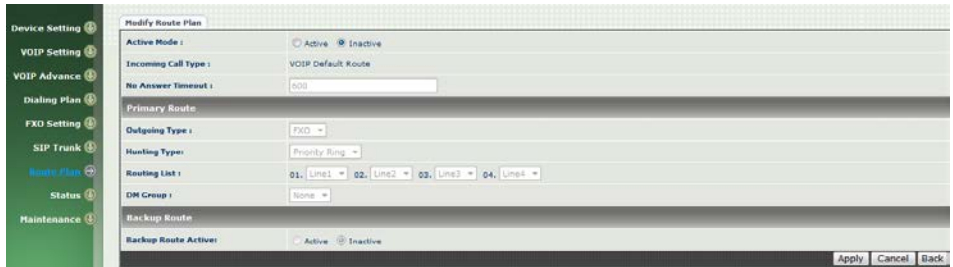


Figure 12.

Set the **Incoming Call Type** to **VOIP**; **Matched Prefix** number from 1001 to 1004; **Matched Length** to **0** and **Outgoing Type** to **FXO**. See Figure 12.
 For example: Configure Route Plan to each line of FXO Gateway.

a. Line #1 of FXO Gateway setting, see Figure 13.

An FXS port 1 incoming call routes to FXO port 1 configuration example.

Incoming Call Type : VOIP

Matched Prefix : 1001

Matched Length : 0

No Answer Timeout : 30 (MUST be longer than 10 seconds).

Outgoing Type : FXO

Hunting Type : selected anyone.

Routing List : 01. Line 1 ; 02. None. ; 03. None ; 04. None.

Routing List is going to select the first priority of Line number 1 to 4. At this application, Only Line 1 is allowed to connect with remote FXS telephone 1.

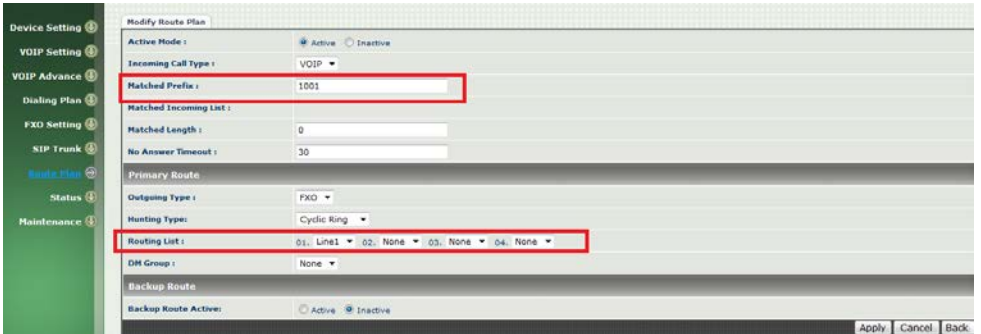


Figure 13.

b. Line #2 of FXO Gateway setting, see Figure 14.

An FXS port 2 incoming call routes to FXO port 2 configuration example.

Incoming Call Type : VOIP

Matched Prefix : 1002

Matched Length : 0

No Answer Timeout : 30 (MUST be longer than 10 seconds).

Outgoing Type : FXO

Hunting Type : selected anyone.

Routing List : 01. Line 2 ; 02. None. ; 03. None ; 04. None.

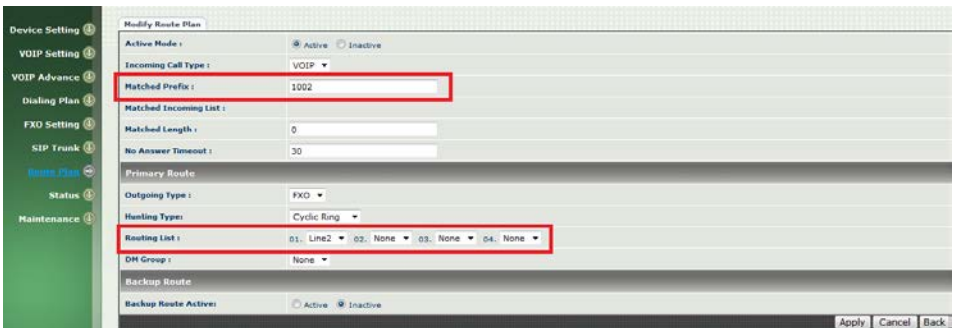


Figure 14.

c. Line #3 of FXO Gateway setting, see Figure 15.

An FXS port 3 incoming call routes to FXO port 3 configuration example.

Incoming Call Type : VOIP

Matched Prefix : 1003

Matched Length : 0

No Answer Timeout : 30 (MUST be longer than 10 seconds).

Outgoing Type : FXO

Hunting Type : selected anyone.

Routing List : 01. Line 3 ; 02. None. ; 03. None ; 04. None.

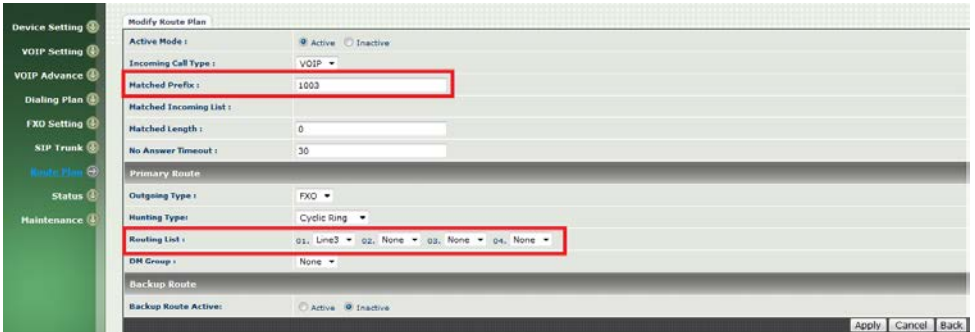


Figure 15.

d. Line #4 of FXO Gateway setting, see Figure 16.

An FXS port 4 incoming call routes to FXO port 4 configuration example.

Incoming Call Type : VOIP

Matched Prefix : 1004

Matched Length : 0

No Answer Timeout : 30 (MUST be longer than 10 seconds).

Outgoing Type : FXO

Hunting Type : selected anyone.

Routing List : 01. Line 4 ; 02. None. ; 03. None ; 04. None.

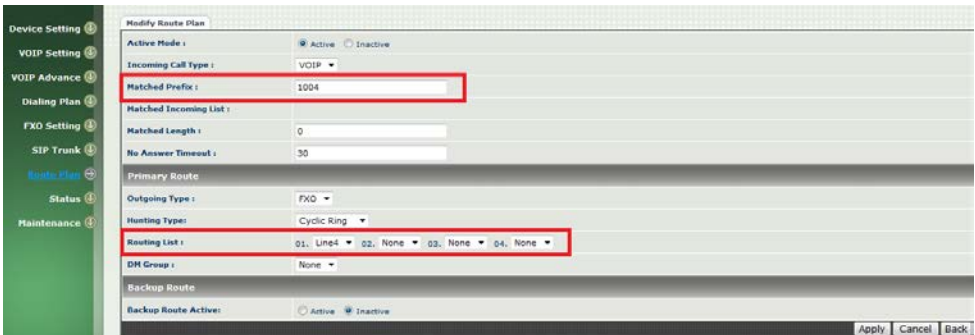


Figure 16.

4. Configure incoming Ring Cadence to FXO Gateway

When you connect WellGate 2540 FXO gateway Lines to local PABX's extension or PSTN switch, please go to WellGate 2540 webpage setting at **VOIP Advance → Ring** to configure proper **Ring Cadence (Ring ON time and Ring OFF time)** according to **the PABX or PSTN ring Cadence specification**.

Without configuring Ring Cadence to FXO gateway properly according to PABX or PSTN Ring Cadence, FXO may detect Ring signal in a strange behavior and cause abnormal operation.

WellGate 2540 factory default Ring Cadence setting is 2 seconds ON, 4 seconds OFF. Some PABX or PSTN provides 1 second ON and 3 seconds OFF Ring Cadence. In this case, please configure WellGate 2540 Ring Cadence to 1 second ON and 3 seconds OFF to match incoming Ring signal. See Figure 17.

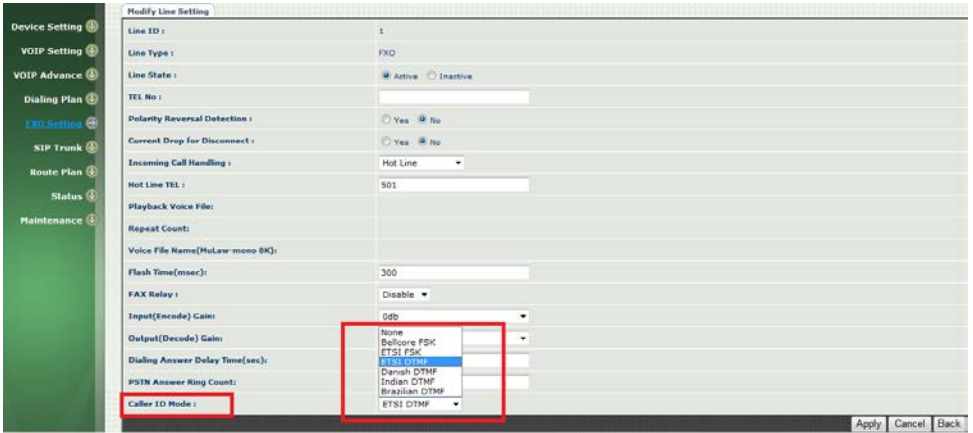


See Figure 17.

5. Enable Caller ID function at FXO and FXS gateway.

The Caller ID configuration at FXO gateway should be configured to match PABX extension or local PSTN Caller ID incoming signal in order to carry this signal to remote FXS gateway to display at FXS analog phone set. Of course, the remote FXS Gateway also has to enable Caller ID generation function (see [Figure 21](#).) Here are steps to enable FXO Caller ID detection specification and enable FXS gateway Caller ID generation steps.

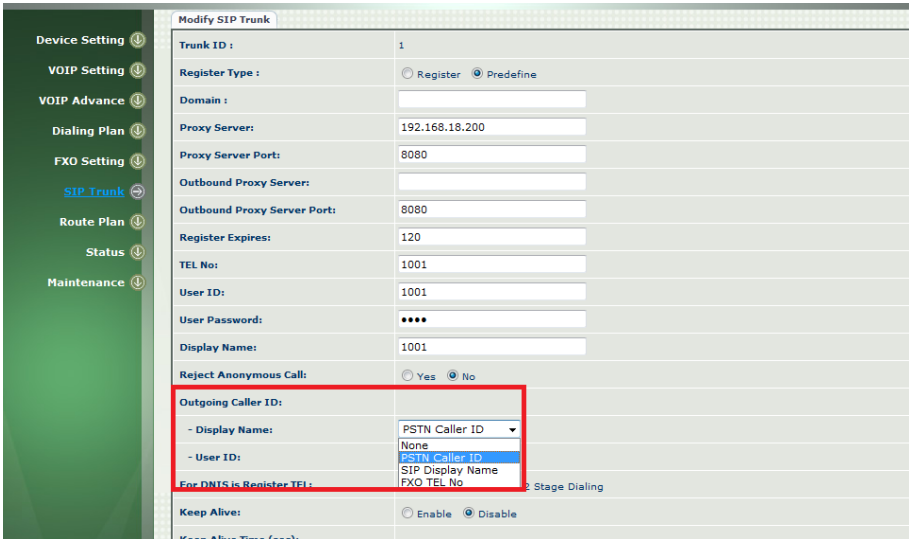
Step 1: Make sure the **Caller ID mode** you have configured to WellGate 2540 FXO gateway is the same as PABX extension or local PSTN switch which WellGate 2540 FXO Line ports have connected. If you don't know what Caller ID mode you need to configure at FXO gateway, please consult to PABX suppliers or PSTN local switch technician to support you. In global market, there are several Caller ID modes which are most used such as **Bellcore FSK**, **ETSI FSK**, **ETSI DTMF**, **Danish DTMF**, **India DTMF**, **Brazilian DTMF**. [See Figure 18](#).



See Figure 18.

Step 2: Go to FXO Gateway webpage configuration at **SIP Trunk → Modify SIP Trunk** to enable **Outgoing Caller ID** with both **Display Name: PSTN Caller ID** and **User ID: PSTN Caller ID** as following [Figure 19](#). After this configuration, FXO gateway send detected Caller ID signal from PABX or PSTN Lines to remote FXS gateway. There are additional two options at **Display Name** and **User ID** to select what number are going to send to remote FXS gateway. Here are summary of three options.

- a. **PSTN Caller ID** : Deliver Line number from PSTN or PABX extension. At this example, they are number from 201 to 205. **See Figure 1**. And FXO Caller ID function should be able to detect Caller ID signal correctly.
- b. **SIP Display Name** : See [Figure 11](#) of four SIP Trunk numbers. At this example, they are number from 1001 to 1004.
- c. **FXO TEL NO.** : See [Figure 20](#) to configure **TEL NO** from **FXO Setting → FXO Line → Modify Line Setting → TEL NO**. Here, default value is **blank**. This option allows you to configure your desire Line number to deliver to remote FXS gateway.



See figure 19. FXO Gateway Caller ID Mode.

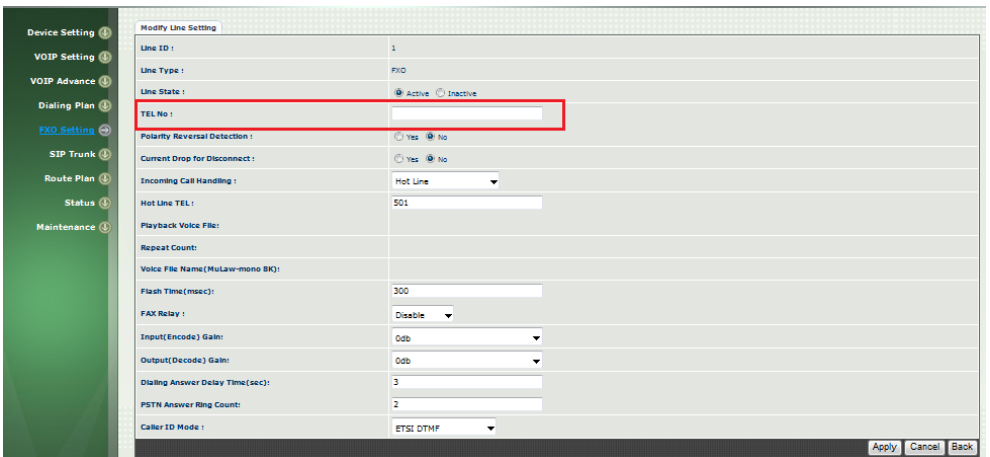


Figure 20. FXO Telephone Number.

Step 3: Enable Caller ID generation at WellGate 2504 FXS gateway. See [Figure 21](#).

Go to **FXS Setting** → **FXS Line** → **Caller ID Mode** to select **Transparent** to enable Caller ID generation. After enable this command, analog telephone set is able to display Caller ID either **FXS phone number** (at this example, they are 501 to 504) or display number which were sent from FXO gateway (**PSTN, SIP Trunk or FXO Line number**).

Step 4: Select FXS gateway's Caller ID Mode. See Figure 22 and 23.
 FXS Gateway Caller ID generation is able to configure which Caller ID mode in order to match analog phone set's Caller ID specification. Go to **FXS Setting** → **Caller ID** → **Caller ID Mode** to select one of Caller ID Mode (**DTMF**, **FSK Bellcore**, **FSK ETSI**, See Figure 23). There are other options you may need to pay attention at **Polarity Reverse Before Caller ID**, **Dual Tone Before Caller ID**, **Caller ID Present**, **DTMF Caller ID start and stop digit**. See Figure 22.

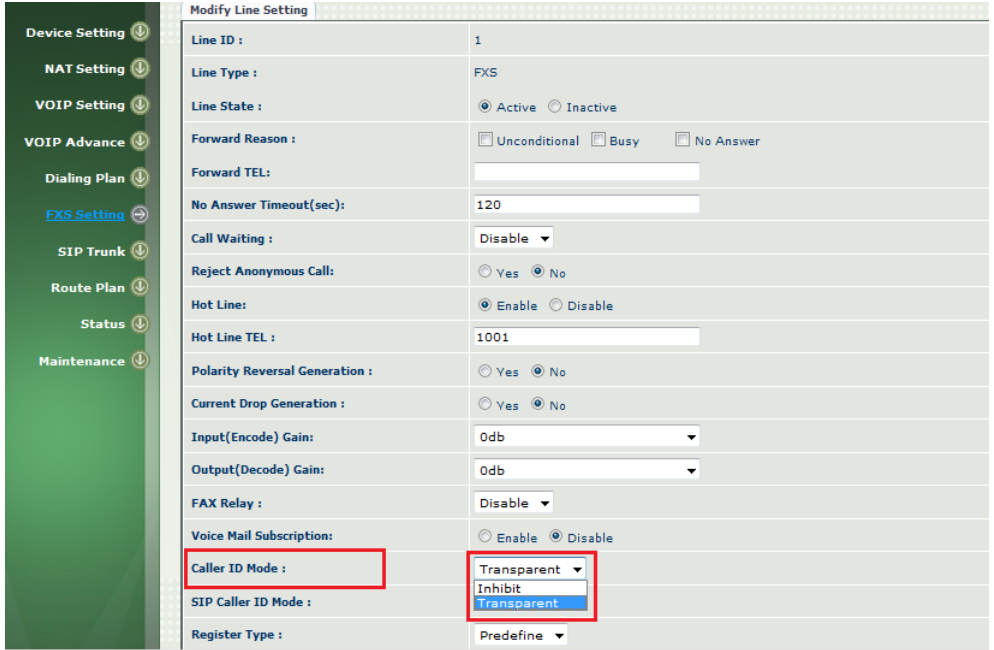


Figure 21. Caller ID feature enable.

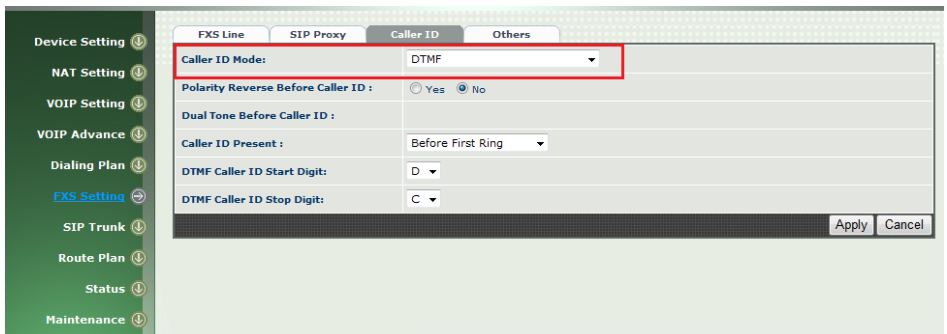


Figure 22. Select Caller ID Mode.

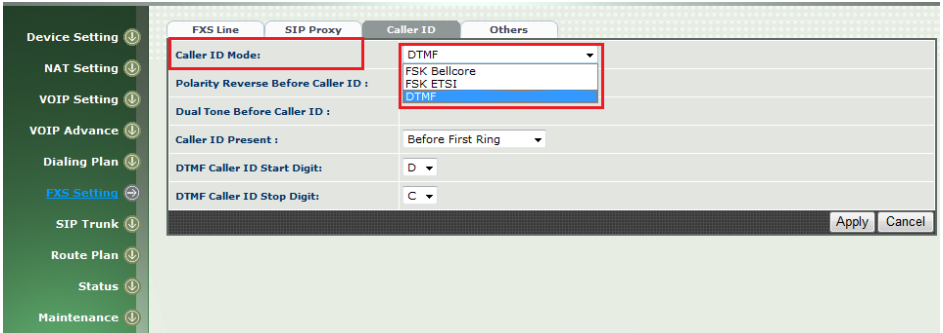


Figure 23. Three options of Caller ID Mode.

6. Release FXO Line from PSTN or PABX extension after call was dropped.

FXO analog interface detect disconnect tone which was sent from PABX extension or PSTN line to release tip/ring phone wire. In other word, the FXO line can only drop call after detect an effective disconnect tone. In FXO gateway webpage configuration, Go to **VOIP Setting** → **Tone** to configure **Disconnect 1 and Disconnect 2** in which disconnect tone are needed to match with PABX or PSTN disconnect tone. See figure 24. For details description, please refer to User manual for 2504 and 2540 firmware ver 2.01 at page 31 in details. You can also go to **Country Template** to select default setting by each country.

Tone \ Setting	Signal Type	Freq 1 (0-3000 Hz)	Freq 2 (0-3000 Hz)	Level 1 (0-99dB)	Level 2 (0-99dB)	On 1 (0-1920ms)	Off 1 (0-1920ms)	On 2 (0-1920ms)	Off 2 (0-1920ms)	Deviation (0-20)
Dial	Continuous	350	440	13	13	500	0	0	0	10
Stutter Dial	Cadence	350	440	13	13	1000	100	0	0	10
Ring Back	Cadence	440	480	13	13	1000	2000	0	0	10
Busy	Cadence	480	620	13	13	500	500	0	0	10
Call Waiting	Cadence	330	440	13	13	280	280	280	0	10
RDH	Continuous	1400	1750	13	13	10000	0	0	0	10
Warning	Cadence	900	0	13	13	500	0	0	0	10
Holding	Cadence	900	0	13	13	500	500	0	0	10
Disconnect 1	Cadence	480	620	13	13	500	500	0	0	10
Disconnect 2	Cadence	480	620	13	13	280	280	0	0	10

Figure 24. Disconnect tone table.

WellGate 2540 supports an alternative feature to release FXO port by detection Loop Current Drop signal. Go to **FXO Setting** → **Select FXO Line** → **Modify Line Setting** → **Current Drop for Disconnect** to **YES** to enable this feature. Once calling user drop this call, PSTN or PABX may **drop Loop Current signal** for about 300 msec (default setting) to inform FXO interface. To detect Loop current drop duration can be changed. Go to **FXS Setting** → **Others** → **Current Drop Time**. See Figure 26. This feature depends on the PSTN switch or PABX you are using. See Figure 25 at webpage configuration.

Modify Line Setting

Line ID :	1
Line Type :	FXO
Line State :	<input checked="" type="radio"/> Active <input type="radio"/> Inactive
TEL No :	
Polarity Reversal Detection :	<input type="radio"/> Yes <input checked="" type="radio"/> No
Current Drop for Disconnect :	<input checked="" type="radio"/> Yes <input type="radio"/> No
Incoming Call Handling :	Hot Line ▾
Hot Line TEL :	501
Playback Voice File :	
Repeat Count :	
Voice File Name(MuLaw-mono 8K):	
Flash Time(msec):	300
FAX Relay :	Disable ▾
Input(Encode) Gain:	0db ▾
Output(Decode) Gain:	0db ▾
Dialing Answer Delay Time(sec):	3
PSTN Answer Ring Count:	2

Figure 25. Loop Current Drop to release FXO port

FXS Line	SIP Proxy	Caller ID	Others
Min Flash Time(80~800msec):		400	
Max Flash Time(80~800msec):		800	
Current Drop Time(msec):		300	

Apply Cancel

Figure 26. Loop Current Drop detection Time.

7. FLASH command generation and detection on FXO and FXS gateway.

FLASH feature is widely used for PABX extension to implement PABX's advanced features such as **Call Transfer, Call Hold, Conference, Forwarding or Call Park and others**. PABX relies on extension's analog phone set to **generate Flash time** to inform PABX this is an **FLASH command**.

In this FXO+FXS P2P Hotline application, FXS needs to detect a FLASH command from analog phone set and transmit to FXO gateway via VoIP network. Then, FXO generates an equivalent Flash time to PABX. Here are step to step guidance to configure this function at both FXO and FXS gateway.

Step 1 : Setup FXS Flash detection time range.

Since FLASH command was initialized from analog phone set, check what flash time this phone set can generate. Then, configure FXS gateway flash time detection range. See [Figure 27](#) as follows. Go to **FXS Setting → Others → setup Minimum and Maximum Flash Time range**. **Note : the analog phone flash time should be located inside FXS gateway setting range to detect an Flash command correctly**. Once FXS detect this command, it sends to FXO gateway via SIP IP protocol automatically.

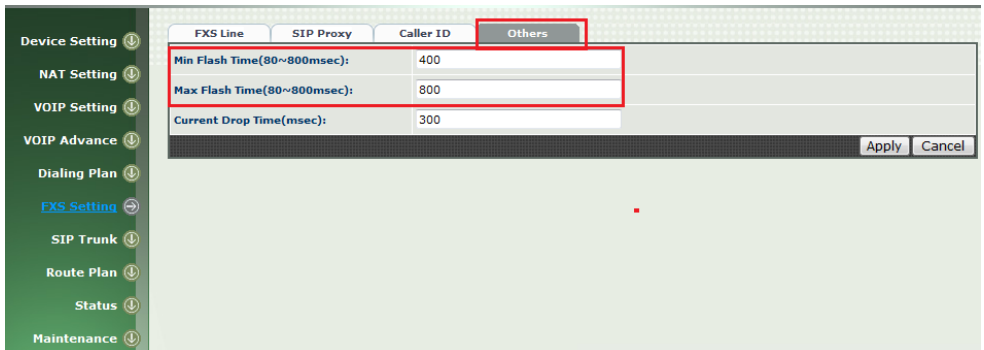


Figure 27. Setup FXS Flash detection time range.

Step 2 : Setup FXO Flash time generation.

Go to FXO Setting → FXO Line → Modify Line Setting → Flash Time (msec). See [Figure 28](#). Flash time has to match PSTN or PABX Flash command specification. Consult to PSTN or PABX technician to fill in proper time.

Modify Line Setting	
Line ID :	1
Line Type :	FXO
Line State :	<input checked="" type="radio"/> Active <input type="radio"/> Inactive
TEL No :	<input type="text"/>
Polarity Reversal Detection :	<input type="radio"/> Yes <input checked="" type="radio"/> No
Current Drop for Disconnect :	<input checked="" type="radio"/> Yes <input type="radio"/> No
Incoming Call Handling :	Hot Line ▾
Hot Line TEL :	501 <input type="text"/>
Playback Voice File:	
Repeat Count:	
Voice File Name(MuLaw-mono 8K):	
Flash Time(msec):	300 <input type="text"/>
FAX Relay :	Disable ▾
Input(Encode) Gain:	0db ▾
Output(Decode) Gain:	0db ▾
Dialing Answer Delay Time(sec):	3 <input type="text"/>
PSTN Answer Ring Count:	2 <input type="text"/>

Figure 28. Setup FXO Flash time generation.