

Release Notes

7800VDPX & 7800VDOX

2.23 – 19th December 2012

Enhancements:

- 1> Support for 3G/LTE dongles added
- 2> Support for static ARP added
- 3> Added 25 & 30 sec to Greeting Delay selection list
- 4> Support for Authentication ID on the SIP account page. (For some specific ISP's)
- 5> Added Option66 to the DHCP server page. It is used for some Set top Boxes that need to get the required information from specified TFTP server.
- 6> Support for Editing on DDNS page.
- 7> Support for WLAN with Bandwidth Usage.
- 8> Support for various DMTF methods -- RFC2833, Inband, SIP Info, SIP Info(short) and Auto. The default is RFC2833.
- 9> Added Time Schedule setting to URL page.
- 10> Changed the default VoIP configuration -- only one SIP account (SIP1) and the VoIP dial plan [X.@SIP1](#) added for phone1 and phone2.
- 11> Marked the port for Default Gateway on summary page. User can easily see which WAN profile the default traffic is being routed.

Bugs Fixed:

- 1> Reduced PSTN echo and noise impact.
- 2> Fixed issue whereby the VoIP dial plan is not updated if Fax Recipient's path is changed.
- 3> Disable T.38 with specified account if this A/M is active.
- 4> Show the latest date first with the Call Log.
- 5> Hide the password on VoIP page.
- 6> Fixed issue whereby VPN tunnel might be built on unexpected WAN interface.
- 7> Resolved system crash when WAN IPv6 is UP and no IPv6 IPsec is configured.
- 8> Fixed issue whereby the IPv6 RADVD packet content is not correct. It might cause MAC's with old OS to crash.
- 9> Fixed issue whereby Bandwidth Usage can't be displayed with IE10.
- 10> Fixed issue whereby WLAN information is not shown with SNMP.
- 11> Fixed issue whereby secondary DNS is not effective after manually changed.
- 12> Fixed issue whereby URL page can't be submitted if there is an entry on the Timer Schedule page.
- 13> Fixed issue whereby outgoing VoIP call might cause VoIP to fail (ISP replies with 183 instead of 180); this is specifically for PennyTel ISP.

- 14> Removed the SIP Registration successful from the syslog and added the failure case to the syslog. This prevents syslog from filling with SIP Registration information.
- 15> Fixed issue whereby VoIP status page might not sync in real time.
- 16> Fixed issue whereby the IPv6 ICMP Filtering rule didn't work.
- 17> Add protection/limitation to prevent SIP/PSTN call from disconnecting abnormally.
- 18> Fixed issue whereby VoIP incoming calls won't work if the port is not 5060.
- 19> Adjusted Call Log content to correctly show the display name.

Known issues:

- 1> The date is not displayed correctly with the samba service