



# Product Release Notice

## ErisStation® 1.1.1.1 Firmware

Release Date: September 2015

This document contains the release notes for VTech ErisStation version 1.1.1.1. This upgrade release, affecting the VCS754 SIP Conference Phone with Four Wireless Mics, includes significant enhancements; therefore, users of previous ErisStation releases are encouraged to upgrade. The following topics are detailed below.

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### What's new?

With the release of 1.1.1.1, the following features have been added. Numbers in parentheses indicate VTech's internal tracking ticket number.

#### **The LED status indicators can be configured for message waiting and missed calls (#1927)**

- The LED status indicators can also be configured to indicate missed calls. When the `user_pref.notify.led.missed_call.enable` parameter is enabled, the LEDs turn on for missed calls and new messages. When disabled, the LEDs turn on for new messages only.

#### **More control over the Provisioning resync interval (#2058)**

- Provisioning checks can now be scheduled at a specific start time and end time on selected days of the week, using the WebUI or configuration file.



## Recent changes

VCS754:

<i>Description</i>	<i>Reference Number</i>
Corrected bug with local conference audio when using SRTP.	#1971
Deleted non-applicable Headset option from the Default Audio Mode selection on the WebUI.	#1986
Introduced Account Label for identifying line/account instead of using the User ID.	#579
Corrected WebUI Local Directory, Blacklist, and Certificates pages with text wrap issues.	#2020
Packetization time ("ptime") is adjustable using the WebUI and configuration file (using the <code>sip_account.x.preferred_ptime</code> parameter).	#1438, #1974
Apply Call Forward Busy policy if Call Waiting is disabled.	#649
Corrected WebUI directory editing to allow user to clear first or last name without being flagged as an error.	#1634
Corrected condition where audio was interrupted when a call on Account 1 was unsuccessfully transferred to Account 2.	#1723
Admin can now select between allowing or disallowing auto answer to occur during an active call using the new <code>sip_account.x.auto_answer_during_active_call</code> parameter.	#1992
Corrected situation where long calls (over 13 hours) spontaneously hung up.	#1916
Extended in-band DTMF digit generation interval to ensure server can properly and consistently detect the digits in speakerphone mode.	#2021
Due to a clash between settings causing failure to log in to the WebUI using https, the <code>web.http_enable</code> parameter was eliminated in favour of a <code>web.https_enable</code> parameter only.	#2110
Revised how manual time settings are applied using the WebUI.	#1086
Corrected audio loss with first call after forwarding an incoming second call.	#1904
Corrected outgoing anonymous calls from sending different P-Preferred Identify header.	#2039
Prior to this version, only provisioning interval resync mode was available. To maintain this mode of operation (when <code>provisioning.resync_time &gt; 0</code> ) with the new firmware, the new parameter <code>provisioning.schedule_mode</code> is set to <code>interval</code> by default.	#2356
The new missed call feature ( <code>user_pref.notify.led.missed_call.enable</code> ) is now disabled by default.	#2359



## Configuration file parameter changes

The following configuration file parameters have been introduced or modified in 1.1.1.1.

sip_account.x.preferred_ptime	New parameter
sip_account.x.account_label	New parameter
network.eapol.access_password	Renamed parameter*
provisioning.fw_server_access_password	Renamed parameter
provisioning.server_access_password	Renamed parameter
profile.admin.access_password	Renamed parameter
profile.user.access_password	Renamed parameter
sip_account.x.authentication_access_password	Renamed parameter
remoteDir.ldap_access_password	Renamed parameter
remoteDir.broadsoft_access_password	Renamed parameter
sip_account.x.auto_answer_during_active_call	New parameter
provisioning.schedule_mode	New parameter
provisioning.weekdays	New parameter
provisioning.weekdays_start_hr	New parameter
provisioning.weekdays_end_hr	New parameter
user_pref.notify.led.missed_call.enable	New parameter

\* The renamed parameters have been created as part of the cleanup of the password export feature. Please read the following compatibility notes:

1. user/admin can continue to log in with the previous password after firmware upgrade; however, our export configuration file and Administrator and Provisioning Guide will only list the new namespaces starting from the new version to avoid any confusion.
2. All old password parameters remain useable and will function identically to the new password namespaces after upgrade.
3. The two new profile.x.access\_password fields won't be exportable until they have been configured at least once with the new firmware via either the new or old parameters.

## How to upgrade

Please refer to the Administrator and Provisioning Manual for your ErisStation product, available [here](#).

## Version compatibility

Version 1.1.1.1 is compatible with previous versions.

## Manuals

For more information on the features described in this document, please reference the manuals and technical documents located here: <http://businessphones.vtech.com/support/manuals>

## Product support

For additional questions, please contact the VTech Business Phones team. Contact information is available here: <http://businessphones.vtech.com/support/contact-us>