Release Notes
for Kerio Operator 1.0.1

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1 Welcome to Kerio Operator 1.0.1

Kerio Operator is a PBX software for small and medium business customers. The main focus of Operator is ease of use, both for the end users and the administrator. Operator is based on the VoIP technology internally but it also supports the standard telecom interfaces like PRI and EuroISDN.

2 Changes in Kerio Operator 1.0.1 (since 1.0.0)

- + Recorded calls can be downloaded
- - Voicemail number was not associated with the voicemail access key on some auto-provisioned phones
- - Custom voicemail greetings and recorded user name were ignored
- - Snom models 320, 370, 821, and 870 were not correctly provisioned when an extension was assigned manually
- - E-mail address could not be used as user name in SMTP authentication
- - External interfaces always had English as the default language for system messages even if the system default was another language
- - Backup sometimes returned empty file
- - Codecs were sometimes reset to the default order
- - Some Czech ordinals were not correctly inflected by the voicemail interface
- - iLBC and G.722 were not marked as "pass-through only" codecs
- - DST parameters were not correct in auto-provisioned configuration for Cisco 7960
- - The field for external phone numbers in SIP interface was limited to 67 characters
- - Session expired when connecting to Control and Operator administration consoles through the same host at the same time
- - Incoming calls went directly to fallback number with some SIP providers
- - Incorrect CallerID was used for outgoing calls with some SIP providers

3 Changes in Kerio Operator 1.0.0 (since RC2)

- + Added new EULA
- + Added support for sending of crash dumps
- + Added generated default password for provisioned phones
- - Fixed installation from an external CD drive
- - Fixed logging of Asterisk debug messages in the debug log
- - Corrected several translations in the Admin GUI
- - Corrected the splash screen on the HW appliance
- - Corrected a regression in generating number rewriting rules for PRI trunks
- ! SIP interface codec order resets to default when changing incoming call routing rules

4 Changes in Kerio Operator 1.0.0 Release Candidate 2 (since RC1)

- - fixed missing module identification in debug log
- - fixed configuration recovery issue that could cause reboot failure
- - allow configuration recovery only in the same or newer version
- - fixed unnecessary restart of provisioned Linksys phones
- - fixed a bug when exporting large log segments
- - fixed an error in adding an extension to a mapped Active Directory user
- - incoming call was always routed to the fallback number on an interface with a single external number
- - renamed "G.729" to "G.729 pass through only" in codec configuration
- - fixed MWI being lit if there was a configuration error
- - fixed warning about '_X.' in the dial plan
- - fixed saving of long SIP interface names
- - fixed fallback to an external number for ring groups
- - fixed dialing to voicemail pickup from external numbers
- - fixed deleting interfaces with multiple external numbers
- - fixed error when updating non-LDAP user attributes with a read-only AD account
- + added possibility to create a dump of asterisk configuration for support
- + improved performance of the MyPhone interface
- + added option to display SIP debug messages in debug log
- + forwarding an extension to a single number now dials directly instead of using the find-me module
- + added emergency number lists for UK and Germany
- ! SIP interface codec order resets to default when changing incoming call routing rules

5 Changes in Kerio Operator 1.0.0 Release Candidate 1 (since Beta 3)

- - fixed access to voicemail in MyPhone
- - added new Beta EULA
- - fixed connection to the configuration database that occasionally failed
- - fixed missing logging during the upgrade process
- - corrected order of name servers in resolv.conf
- - sending voice mail over SMTP sometimes did not work
- - fixed extensions reported as not registered in MyPhone
- - increased timeout for large uploads
- - corrected Czech translation in MyPhone
- * ssh does not start automatically
- - fixed NTP warning after restart
- - fixed searching for users in Active directory / LDAP
- - fixed testing of directory server connection
- - added missing DTMF transfer types
- - fixed broken Operator-Operator trunk
- - fixed design glitches in Safari 5
- - fixed ext3 warnings after restart
- - fixed vfat warnings after restart
- - fixed transcoding WAV files to 8 kHz
- - fixed music on hold for calls held from SIP phones
- + added German system messages
- + added Kerio's own default music on hold
- + added redirection from HTTP to HTTPS, optional for MyPhone
- ! external numbers do not work as fallback numbers in Ring groups

6 Changes in Kerio Operator 1.0.0 Beta 3 (since Beta 2)

- - Fixed new voicemail indication
- - Fixed branding in VMWare appliance title
- - Corrected displaying of Trial license/Trial registration information in the splash screen
- - Editing of external numbers was not possible if they were similar to outbound route prefixes
- - Fixed registration of several SIP accounts with the same provider
- - Fixed searching for users in Active Directory outside of the standard location
- - User's full name was not visible when initiating call from MyPhone
- - Fixed upgrade issues on very slow virtual machines
- ! System messages in German not yet available

7 Changes in Kerio Operator 1.0.0 Beta 2 (since Beta 1)

- + It is now possible to upload firmware files for provisioned phones
- + Added provisioning for Snom models 300, 360, 820, M3, MeetingPoint
- + Added system messages in American English and British English
- * Improved call history
- * Improved the look & feel of the User console ("MyPhone")
- * Added call history to the User console
- - Fixed searching in logs
- - It was sometimes not possible to save Auto Attendant submenu
- - Incoming call was sometimes terminated on EuroISDN port
- - Packet sniffer data were incorrectly removed after failed download
- - Status of SIP interface was sometimes incorrectly reported
- - Logout from MyPhone sometimes caused logout from Admin console
- - Backup sometimes caused 100% CPU load
- - MWI was activated on all provisioned phones if there was voicemail access problem
- - Provisioning did not work with newer Linksys models
- - Fallback number was not generated for an interface that had it configured
- - BRI interface disappeared from interface list after a reset
- - Active calls grid sometimes showed incorrect data
- - Fast switching between screens reported changes in partially loaded dialogs
- - Admin password containing a white space did not sync with root
- ! System messages in German not yet available
- ! Upgrade from older versions to Beta 2 may not succeed on VMWare Server 2.0 running on 64-bit linux systems. Operator returns to the previous version safely. The recommended solution is to install a new virtual appliance and transfer Operator's configuration using backup and restore.

8 Open Source Software Notice

Kerio Operator includes open source software. The complete open source code packages of these components are available in Kerio Software Archive at http://download.kerio.com/archive/.

9 Legal Notice

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