Webitel 3.0.x release notes

3.0.7

- Core Server
    - Added new API for working with blacklists | #84
    - Added new API for working with the SMTP protocol. Added ability to set SMTP server settings for each domain | #86
    - Added the ability to send a recording of a conversation or a received FAX message to one or several Email addresses| #86
    - Code optimization and improvements in the WebSocket server.
- Advanced Call Router
    - Updated application queue | #65
    - Updated application recordSession - added ability to set file type | #77
    - Added the ability to record video calls | #77
    - Added new application receiveFax to get FAX into PDF file| #78
    - Added new application blackList to perform any actions with numbers that are entered the black list| #78
    - Work has begun on creating a graphical routing designer.
- CDR Server
    - Added new API to delete conversation records| #39
    - Added restriction for users to receive statistics - now users only see calls of their number | #39
    - The server returns the type of file(s) associated with each call. (audio/mp3, video/mp4, application/pdf) | #39
    - Added the ability to upload the registry of calls into the Excel file.
    - Work has begun on closer integration of statistics with Kibana for the ability to build reports.new application
- FreeSWITCH
    - Added scripts to download PDF and MP4 files to CDR Server | #192
    - Fixed bug when there is no network interface with IPv6 address | #190
    - Video Call Improvements | #185

3.0.6

- Core Server
    - Added the ability to get a list of queues in which the agent is included | #64
    - Added new events with information about the number of subscribers in the queue | #64
- Webitel Console
    - Fixed bug with determining registration of WebRTC subscribers by IPv6 | #84
- Advanced Call Router
    - Refactoring and code optimization. Improved performance.
- CDR Server
    - support for elasticsearch and kibana added | #25
    - fixed error with marking logs in logstash | #25
    - new API for getting the size of conversation records by Id or Domain | #29
- FreeSWITCH
    - upgrade to version 1.6 | #45
    - IPv6 support added for SIP and WebRTC | #45
    - added video conferencing support | #45
    - fixed a bug with the lack of voice, if a SIP subscriber without a video calls to a WebRTC subscriber with a video | #175

3.0.5

- Core Server
    - new API interface for connecting to the conversation | core-3.0.5-r14
    - the ability to set the language for sound phrases for the domain | core-3.0.5-r14
- Advanced Call Router
    - added a new application ivr | acr-3.0.5-r3
    - added ability to set and use global variables in routing | acr-3.0.5-r9
    - Added application for working with voicemail | acr-3.0.5-r9
    - expanded functionality of the application for working with call queues | acr-3.0.5-r9
    - IF and SWITCH functions added | acr-3.0.5-r9
- Refactoring the client library code. Added statuses for agents in the call queue.
- Added the ability to install using Ansible.
- Logstash support added.

3.0.4

- Advanced Call Router
    - added time zones and the ability to create ACR Scheme based on the date and time.
    - new applications added: park and queue.
    - new application for playing sound files: local and remote (MP3 or WAV)
- Core Server
    - implemented the ability to work with CDR Server using REST API v2 commands.
    - added new REST API v2 to work with calls.
- CDR Server
    - added the ability to download WAV and MP3 files for further use in ACR Scheme
- The code for working with video calls has been added to the client library.

3.0.3

- Docker images were optimized, which allowed on-site installation on client servers.
- Implemented the first version of the web administrator.