PBXware

version 5.1.2

Changelog

PBXware [5.1.2]

Features:

1. **PJSIP : New option 'Allow IP Address authentication for Extensions'.**
2. **Spanish speech to text.**
3. **French speech to text.**
4. **French translation for GUI.**
5. **New Polycom Firmware (version 5.7).**
6. **TFTP Server : disabled by default for new installations, with the ability to turn it on again.**

Bug fixes:

1. Rotate mon.log files properly.
2. Disabled CDR's for static and callback members.
3. Removed old CDR queries.
4. Voicemail : Fixed transcription request error handling.
5. HTTP : Increased request time for API calls.
6. Queue callback: agent answer mode, use outbound Caller ID from now on.
7. DIDs : QUEUE_PRIO variable channel inheritance.
8. DB : Fixed bug where remotedb was closed before update.
9. Do not Disturb: Ignore temporary DND setting when switching DND.
10. Queue statistics: callback report handle new transfer events.
11. Extension operation time apply before DND.
12. Recordings : Fixed beep, if recording is on B side.
13. Dashboard : Added Push notification status to json feed as a combo of ari-dial and pns statuses.
14. Access codes : Queue interupt message was deprecated.
15. Access code : do not allow agent greeting on MT.
16. DIDs : Disabled Greeting if Closed Operation Times is ON.
17. DIDs : Disabled Recording if destination is fax.
18. LXC : Detect LXC via /proc/1/environ.
19. Auto provisioning : Removed support for Xlite.
20. IVR statistics : Changed log_ivr_calls table date_time field from varchar to int(11).
21. File system : Small fix for cleanup of faxes.
22. IP Restrictions > Policies: Fixed for Edit and Delete buttons.
23. System > Sangoma : Removed BRI cards.
24. PJSIP : Fixed maximum expiration spelling mistake.
25. Multi-Digit IVR / Pin-Based IVR : Fixed Delete action for value '0'.
26. PJSIP : Fixed spelling for option disable_direct_media_on_nat.
27. Extensions : Fixed "Copy as New", resetting Caller ID name.
28. queue : Exit digit now can be "0".
29. Servers : Increased emergency services to 16 digits.
30. Auto provisioning : Fixed BLF fetching on MT for additional MAC addresses.
31. Extensions : "Copy as New" destinations permissions fix.
32. Reports : get_dropdown_trunks can now return name as id, fix for reports page.
33. Reports : Trunks section was completely broken.

34. **Auto provisioning : Fixed custom parameter provisioning on Panasonic UTG models.**

35. **Queues panel : setting "today" data default, refreshing data.**

36. **Fixed Operations Times on IVR not working properly.**

37. **Voicemail Edit : 'old file' download fix.**

38. **Reports : Listen to Recordings volume slider height was broken.**

39. **Monitor > Extensions: Monitor action is broken on MT.**

40. **Ring groups edit - Now preserve destinations order.**

41. **semantic-ui : added 'sorted' class on dropdown.**

42. **DIDs : Fixed CSV upload where number_zerofill is "0".**

43. **Enhanced Services : Moved 'Ext Operation Times' above 'Call Filters & Blocking' in the ES list.**

44. **Conferences : Disabled writing of conferences to meetme.conf.**

45. **Voicemail : Reload action added, when creating new custom Voicemail extension.**

46. **DIDs CSV : "External Number" and "Deny Access", CSV upload was broken.**

47. **Dashboard : Swapped position between Chat and Push Notifications.**

48. **Dashboard : Replaced TFTP status with Push Notifications status.**

49. **Trunks : Hostname fields are stripped of additional spaces.**

50. **Transport settings are now correctly set on extension according to UAD setup.**

51. **CDR: Fixed bug where multiple files with the same name can be present in the same zip archive.**

52. **Extensions : Fixed Caller ID Name not escaping HTML quotes.**

53. **Settings > About : Fixed Licensee details in case Branding is on.**

54. **Reports > CLI messages : Restored Google search from v4.**

55. **PJSIP : Now saving additional mailboxes properly.**

56. **PJSIP : Disabled write of 'disallow' field per endpoint.**

57. **PJSIP Trunks : Outbound Proxy added sip: prefix.**

58. **Branding : restricted to Site Administrators only.**

59. **Search : Fixed when there are special characters in search query.**

60. **Server Details : Hostname needs to be validated.**

61. **SSL Certifications : Now enabled if MySQL is down.**

62. **Revert "res_musiconhold : Start playlist after initial announcement".**

63. **Woomera devices completely removed.**

Bicom Systems (USA)
3901 South Ocean Drive
Suite 9E
Hollywood, Florida
33019-3003
United States

Tel: +1 (954) 278 8470
Tel: +1 (619) 670 7777
Fax: +1 (954) 278 8471

sales@bicomsystems.com

Bicom Systems (CAN)
126C Hampton Rd
Suite# 316
Rothesay, New Brunswick
E2E 2N6
Canada

Tel: +1 (647) 313 1515
Tel: +1 (506) 635 1135

sales@bicomsystems.com

Bicom Systems (UK)
Unit 5 Rockware BC
5 Rockware Avenue
Greenford, London
UB6 0AA
United Kingdom

Tel: +44 (0) 20 33 99 88 00
Fax: +44 (0) 20 33 99 88 01

sales@bicomsystems.com

Bicom Systems (France)
188 Route de Blessy
St. Quentin
Aire-sur-la-Lys
62120
France

Tel: +33 (0) 3 60 85 08 56

sales@bicomsystems.com

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