

Avaya J100 Series SIP Release 4.1.9.0 Readme

This file is the Readme for the Avaya J100 Series SIP Release 4.1.9.0 Feature Pack software (J100 SIP 4.1.9.0). This file describes the contents of the July 2025 (**4.1.9.0.8**) release software distribution package.

J100 SIP 4.1.9.0 software is supported on the Avaya J129, J139, J159, J179 and J189 IP Phones used with Avaya Aura[®], Avaya IP Office[™], and select OpenSIP platforms. J100 SIP 4.1.9.0 software will not load or operate on any other models.

This release supersedes all previous Avaya J100 Series SIP software releases. Avaya recommends that all customers using Avaya J100 Series SIP software upgrade to this version at their earliest convenience.

The information in this document is accurate as of the issue date and subject to change.



Please refer to the Advisements in this file for important information prior to deploying this software.

Compatibility

The Avaya J129, J139, J159, J179 and J189 IP Phones using J100 SIP 4.1.9.0 software are supported with:

- Avaya Aura[®] Platform 10.1.0.0 (Avaya Aura[®] Communication Manager 10.1.0.0, Avaya Aura[®] Session Manager 10.1.0.0, Avaya Aura[®] System Manager 10.1.0.0, Avaya Aura[®] Presence Services 10.1.0.0) and associated feature/service packs
- Avaya Aura[®] Platform 10.2.x (Avaya Aura[®] Communication Manager 10.2.x, Avaya Aura[®] Session Manager 10.2.x, Avaya Aura[®] System Manager 10.2.x, Avaya Aura[®] Presence Services 10.2.x) and associated feature/service packs
- IP Office[™] 11.0 or later for J129/J179
 - Refer to IP Office documentation for specific compatibility.
- IP Office[™] 11.0 SP1 or later for J129/J139/J179
 - IP Office[™] 11.0 FP4 or later for support of Bluetooth on J179
 - Refer to IP Office documentation for specific compatibility.
- IP Office[™] 11.0 FP4 SP2 or later for J159
 - Refer to IP Office documentation for specific compatibility.
- IP Office[™] 11.1 FP1 or later for J189
 - Refer to IP Office documentation for specific compatibility.
- IP Office[™] 12.1 or later for IPv6
- Open SIP Platforms
 - Broadsoft Broadworks R22.0
 - Asterisk R16
 - FreeSWITCH 1.8.5
 - Netsapiens v41.2.2
 - Metaswitch CFS V9.5
- Avaya Cloud Office by Ring Central™

Refer to $\frac{\text{https://secureservices.avaya.com/compatibility-}}{\text{matrix/menus/product.xhtml?name=J100+-+SIP\&solution=false\&version=4.1}}$

for an up-to-date listing of compatible products.

New Support in J100 SIP 4.1.9.0

The following new features are included in 4.1.9.0 Feature Pack:

Theme of this Service Pack	Description
Prevent use of MD5 in J100	J100 SIP phone feature to block use of MD5 as described in RFC-8760.
	The spirit of this RFC is to improve security for SIP connections by removing the less secure MD5 digest authentication method.
	The phone already supports SHA-256 and SHA-512 SIP digest authentication.
	The phone will select the first supported digest authentication method from the list of methods provided by the server so disabling MD5 in the phone will ensure it is not used.
J100 SIP/ auto-call back displayed interactively as a softkey	Automatic Callback (ACB) allows internal users who place calls to busy or unanswered internal deskphones to be called back when the called deskphone becomes available.
	If the phone has a lot of features configured, then activating the feature during the ringback state requires scrolling down the features menu quickly to find ACB, if you are too slow the call may have already been terminated or forwarded.
	This feature extends the default softkey set for outbound in progress call to provide "Auto-Callback" feature as a soft key
J100 SIP Guest Login with "Groups"	Groups can be set via PhoneUI or SMGR/PPM and causes a phone reboot to pick up group specific settings.
	Until now a Guest Login was not supported when Groups were set via SMGR since the phone has to reboot and when this occurred the primary user is logged back into the phone.
	In this feature pack J100 now optionally supports a Guest user login when Groups are set from SMGR.
	When this setting is activated the phone will continue to log into the guest user account when the phone is rebooted until the guest user logs out or the guest user timout has expired.
New Phone Menu Option to Set the KEEP_CURRENT_CA	KEEP_CURRENT_CA is an existing setting that was only configurable by the administrator from the settings file.
Option	With this option the end user now has the choice to turn this feature on and off depending on their current preference.
	Supported in Aura, ACO and 3PCC environments. (not IPO)

Theme of this Service Pack	Description
Embed all language files in firmware	J129-J189 model firmware will embed the language files in /AvayaDir
	Firmware zip package will continue to include language .xml files as some customers modify them
	SET LANGUAGES is still used to define which languages will appear for user selection
	If SET LANGUAGES language_filename.xml already exists in /AvayaDir then GET IF-MODIFIED will NOT be performed.
	Any customers that manually modify language_filename.xml will need to save as a different filename.
	Note:
	Previously a provided language file could be modified and used without changing the file name.
	A custom language file now needs to have a different name than any embedded language file name.
ACO: Call Recording from a Feature button	Provides a similar user experience with J100 SIP Phone registered with ACO as with Avaya Cloud Phone application.
	Allows the user to toggle status On/Off by pressing the Call Recording feature button
	Provides J100 SIP User a Call Record feature button that shows current status of Call Record on the active call session

Documentation for J100 SIP 4.1.9.0

The following documentation has been updated for this release:

- Quick Reference for Avaya J129 SIP IP Phones in Avaya Aura®
- Quick Reference for Avaya J139 SIP IP Phones in Avaya Aura®
- Quick Reference for Avaya J159 SIP IP Phones in Avaya Aura®
- Quick Reference for Avaya J169 and J179 SIP IP Phones in Avaya Aura®
- Quick Reference for Avaya J189 SIP IP Phones in Avaya Aura®
- Using Avaya J129 SIP IP Phones in Avaya Aura®
- Using Avaya J139 SIP IP Phones in Avaya Aura®
- Using Avaya J159 SIP IP Phones in Avaya Aura®
- Using Avaya J169 and J179 SIP IP Phones in Avaya Aura®
- <u>Using Avaya J189 SIP IP Phones in Avaya Aura®</u>
- Using Avaya J100 Series IP Phones for Call Center Agents
- Using Avaya J100 Expansion Module for SIP
- Avaya J100 Series IP Phone Overview and Specifications
- Installing and Administering Avaya J100 Series SIP IP Phones in Avaya Aura®

The following documentation has not been updated for this release:

- <u>Using Avaya J129 SIP IP Phone in Open SIP</u>
- Using Avaya J139 SIP IP Phone in Open SIP
- Using Avaya J159 SIP IP Phone in Open SIP
- Using Avaya J169 and J179 SIP IP Phones in Open SIP
- Using Avaya J189 SIP IP Phone in Open SIP
- Installing and Administering Avaya J100 Series SIP IP Phones in Open SIP
- Quick Reference for Avaya J129 SIP IP Phone in Open SIP
- Quick Reference for Avaya J139 SIP IP Phone in Open SIP
- Quick Reference for Avaya J159 SIP IP Phone in Open SIP
- Quick Reference for Avaya J169 and J179 SIP IP Phones in Open SIP
- Quick Reference for Avaya J189 SIP IP Phone in Open SIP

The following Partner Configuration guides are included below for reference:

- Broadsoft Partner Configuration Guide J100 Series
- Asterisk Partner Configuration Guide J100 Series
- FreeSWITCH Partner Configuration Guide J100 Series

These documents are available on http://support.avaya.com under "J100 Series IP Phones" -> "SIP 4.1.x" -> "Product Documents". They are also available on https://documentation.avaya.com under "Phones and Devices" -> "Avaya J100 Series IP Phones".

J100 SIP 4.1.9.0 (4.1.9.0.8) Package Content

The J100 SIP 4.1.9.0 package (J100-IPT-SIP-R4_1_9_0-071525.zip) contains all the files necessary to upgrade Avaya new or previously installed Avaya J129/J139/J159/J169/J179/J189 IP Phones to the J100 SIP 4.1.9.0 software.

Note: There is a second J189 Hardware version that requires a seperate FW image. For more information please see <u>Appendix 1</u>.

- FW_S_J129_R4_1_9_0_8.bin application binary file for J129
- FW_S_J139_R4_1_9_0_8.bin application binary file for J139
- FW_S_J159_R4_1_9_0_8.bin application binary file for J159
- FW_S_J169_R4_1_9_0_8.bin application binary file for J169
- FW_S_J179_R4_1_9_0_8.bin application binary file for J179
- FW_S_J189_R4_1_9_0_8.bin application binary file for J189
- FW S J189A R4 1 9 0 8.bin application binary file for J189
- FW_JEM24_R1_0_1_0_27.bin application binary file for the JEM24
- J100Supgrade.txt This file is downloaded by the IP Phones and instructs the phone on how to upgrade to this version of software.
- Predefined language files for phone display:
 - o Mlf J129 BrazilianPortuguese.xml
 - Mlf J129 CanadianFrench.xml
 - Mlf J129 CastilianSpanish.xml
 - o Mlf_J129_Chinese.xml
 - o Mlf J129 Dutch.xml
 - o Mlf J129 English.xml
 - o Mlf_J129_German.xml
 - Mlf J129 Hebrew.xml
 - Mlf_J129_Italian.xml
 - Mlf J129 Japanese.xml
 - o Mlf J129 Korean.xml
 - Mlf J129 LatinAmericanSpanish.xml
 - Mlf J129 ParisianFrench.xml
 - Mlf_J129_Polish.xml
 - Mlf J129 Russian.xml
 - o Mlf J129 Turkish.xml
 - o Mlf_J139_Arabic.xml
 - Mlf J139 BrazilianPortuguese.xml
 - o Mlf_J139_CanadianFrench.xml
 - Mlf J139 CastilianSpanish.xml
 - Mlf J139 Chinese.xml
 - Mlf J139 Dutch.xml
 - Mlf J139 English.xml
 - Mlf_J139_German.xml
 - Mlf J139 Hebrew.xml
 - Mlf_J139_Italian.xml
 - Mlf_J139_Japanese.xml
 - Mlf J139 Korean.xml
 - Mlf_J139_LatinAmericanSpanish.xml
 - Mlf_J139_ParisianFrench.xml
 - o Mlf J139 Polish.xml
 - o Mlf J139 Russian.xml
 - o Mlf J139 Thai.xml

- Mlf_J139_Traditional_Chinese.xml
- o Mlf J139 Turkish.xml
- Mlf J159 Arabic.xml
- o Mlf_J159_BrazilianPortuguese.xml
- Mlf J159 CanadianFrench.xml
- Mlf_J159_CastilianSpanish.xml
- Mlf_J159_Chinese.xml
- o Mlf_J159_Dutch.xml
- Mlf J159 English.xml
- Mlf_J159_German.xml
- Mlf_J159_Hebrew.xml
- Mlf_J159_Italian.xml
- Mlf_J159_Japanese.xml
- Mlf_J159_Korean.xml
- Mlf J159 LatinAmericanSpanish.xml
- Mlf_J159_ParisianFrench.xml
- Mlf_J159_Polish.xml
- o Mlf_J159_Russian.xml
- o Mlf J159 Thai.xml
- o Mlf_J159_Traditional_Chinese.xml
- Mlf_J159_Turkish.xml
- Mlf J169 J179 Arabic.xml
- o Mlf_J169_J179_BrazilianPortuguese.xml
- Mlf_J169_J179_CanadianFrench.xml
- Mlf_J169_J179_CastilianSpanish.xml
- o Mlf_J169_J179_Chinese.xml
- o Mlf J169 J179 Dutch.xml
- Mlf_J169_J179_English.xml
- o Mlf J169 J179 German.xml
- o Mlf_J169_J179_Hebrew.xml
- o Mlf_J169_J179_Italian.xml
- Mlf J169 J179 Japanese.xml
- Mlf J169 J179 Korean.xml
- Mlf_J169_J179_LatinAmericanSpanish.xml
- Mlf_J169_J179_ParisianFrench.xml
- Mlf_J169_J179_Polish.xml
- o Mlf_J169_J179_Russian.xml
- Mlf J169 J179 Thai.xml
- Mlf_J169_J179_Traditional_Chinese.xml
- o Mlf_J169_J179_Turkish.xml
- Mlf_J189_Arabic.xml
- Mlf_J189_BrazilianPortuguese.xml
- Mlf_J189_CanadianFrench.xml
- o Mlf_J189_CastilianSpanish.xml
- o Mlf J189 Chinese.xml
- o Mlf J189 Dutch.xml
- o Mlf_J189_English.xml
- Mlf J189 German.xml
- Mlf J189_German.xml
 Mlf J189 Hebrew.xml
- Mlf J189 Italian.xml
- o Mlf_J189_Japanese.xml
- Mlf J189 Korean.xml
- Mlf_J189_LatinAmericanSpanish.xml

- Mlf_J189_ParisianFrench.xml
- o Mlf J189 Polish.xml
- Mlf_J189_Russian.xml
- o Mlf_J189_Thai.xml
- Mlf_J189_Traditional_Chinese.xml
- Mlf J189 Turkish.xml
- Eight extended Korean ring tone files:
 - KoreanRT1.xml
 - KoreanRT2.xml
 - KoreanRT3.xml
 - KoreanRT4.xml
 - KoreanRT5.xml
 - KoreanRT6.xml
 - KoreanRT7.xml
 - KoreanRT8.xml
- AvayaLanguageTool_SIP.xlsm Excel tool for creating additional language files
- One certificate file:
 - av_prca_pem_2033.txt Avaya Product Root CA certificate with an expiration date of 2033
- Avaya-J100IpPhone-MIB.mib mib file
- release.xml
- A "signatures" subdirectory containing signature files and a certificate file. Both SHA-1 and SHA-256 signature files are included
- Avaya Global Software License Terms Jun-2024.pdf

System specific parameters should be entered into the 46xxsettings.txt file which is available for separate download at http://support.avaya.com. New/changed configuration parameters with this release of software are shown in

Appendix 3.

Advisements with J100 SIP 4.1.9 software

3PCC Hardware - cannot be used with Avaya Aura® or Avaya IP Office™



Customers who purchased "3PCC" versions of the J129/J139/J159/J169/J179/J189 hardware which are pre-configured for interworking with Open SIP. When using J100 3.0.0.1 or later software, the "3PCC" hardware cannot be converted for use on Avaya Aura[®] or Avaya IP Office™.

J179 with Expansion Modules (JEM24) - 5-volt power supply may be required



There are certain power requirements when connecting the JEM24 expansion modules to the phone. Depending upon the amount of power supplied by the power source over Ethernet, it may be necessary to power the phone by a separate 5 Volt power supply. Please see the Power Specifications section in the "Installing and Administering Avaya J100 IP Phones".

J189 with Expansion Modules (JEM24)



There are specific power requirements when connecting the JEM24 expansion modules to the phone. Power can be supplied by PoE or by a 5V power adapter. The J189 has a physical switch to set PoE power level. The two settings are high (H) and low (L).

When the phone is powered using a 5V power adapter you can connect up to 2 JEM24 expansion modules.

When the phone is powered using PoE then the physical PoE switch on the back of the phone must be set to "H" in order to connect 1 or 2 JEM24 expansion modules.

NOTE: The PoE power level switch should only be changed when the phone is not running.

Public Certificates

J100 SIP software includes 64 built-in public CA certificates from a wide range of vendors which can be used instead of having to explicitly add them via a TRUSTCERTS parameter. The use of these certificates is controlled by the ENABLE_PUBLIC_CA_CERTS parameter. A full list of the certificates is included in Appendix B of "Installing and Administering Avaya J100 IP Phones" and "Installing and Administering Avaya J100 Series IP Phones in an Open SIP environment".

J129/J139/J159/J179/J189 IP Phones - Minimum software release

Avaya periodically releases new hardware variations of the J100-Series IP Phones typically to address a need to change hardware components. That change may require a new version of software to support the new hardware. This then forces a minimum software release supported on that hardware. **Attempts to downgrade these models to lower versions of software will be rejected**. Refer to Appendix 1.

Adding Feature buttons on the Phone Screen in Avaya Aura™

When deploying a J100 IP Phone on Avaya Aura®, there are three ways to make feature buttons appear on the Phone Screen:

- If using Avaya Aura® 7.1.3.3 or later, tag the feature as a "Favorite" in SMGR. This will place the Feature button on the specified key on the Phone Screen
- Features not tagged as "Favorite" in SMGR can be placed on the Phone Screen using the SET PHONEKEY parameter in the 46xxsettings.txt file (this can be applied generically to all phones, on a group basis, or on a MAC-specific basis)
- Using the Phone Key Customization feature the end user can add a Feature key to the Phone Screen (Settings -> Phone -> Phone Keys Customization)

Language Localization Software Tool and Localized Language Files

Avaya includes a Language Localization Tool (AvayaLanguageTool_SIP.xlsm) as part of the software download package. This tool allows users to create custom downloadable language files for the J100-series in addition to the built-in language files.

Additional information on the tool as well as already-made localized language files can be found at https://support.avaya.com/downloads/downloads/downloads/downloads/downloads/contentId=C2019925105008420 7&productId=P1661

Limitations with IPv6

The following are known limitations of the J100 SIP 4.1.0 or later implementation:

- Open SIP Interoperability
- Extended rebind
- LLDP configuration of IPv6 related settings is not supported
- Microsoft Exchange integration over IPv6 must use an FQDN for EXCHANGE_SERVER_LIST. i.e. SET EXCHANGE_SERVER_LIST exch1.myco.com
- Wifi
- The following functionality is only supported via IPv4
 - o Push
 - Avaya Diagnostic Server (ADS / SLAMon)
 - Shared Control / Deskphone Mode
 - o Interworking with CC Elite.

SSH - Remote Access (EASG)

J100 SIP software contains an SSH server which is used only by Avaya Services for debugging purposes. The SSH server supports only Avaya Services Logins ("craft" and

"sroot"). By enabling Avaya Services Logins, you are granting Avaya access to your system. This is required to maximize the performance and value of your Avaya support entitlements by allowing Avaya to resolve product issues in a timely manner. By disabling Avaya Services Logins, you are preventing Avaya access to your system. This is not recommended as it can impact Avaya's ability to provide support for the product. Unless the customer is well versed in managing the product themselves, Avaya Services Logins should not be disabled. The access to the SSH server is protected by EASG (Enhanced Access Security Gateway).

Support for SHA2-signed software files

The software files are signed using SHA-256 digital signatures. J100 SIP software is only capable SHA-256 digital signature verification.

Utility Server 7.1 is the minimum software version required to install J100 4.1.9.0 zip packages

Earlier versions of the Utility Server are not able to install J100 SIP 4.1.9.0 software packages.

Support for OCSP

J100 SIP software supports OCSP (Online Certificate Status Protocol) for checking whether certificates presented to the phone by servers are good, revoked, or unknown. If a certificate is revoked, the TLS connection will not be established or will be closed (in the case of an ongoing TLS connection). OCSP is supported for 802.1x (EAP-TLS), SIP over TLS, WiFi (EAP-TLS) and HTTPS.

MLPP - Limitations during a server failure

Call override/preemption is not available during a preserved call caused by inability to access Session Manager.

Bi-Directional EHS – Compatible Headsets

Compatibility testing of the Bi-Directional EHS functionality with headsets from 3rd-party vendors is undertaken through the Avaya DevConnect program.

Microsoft Exchange Integration using EWS

If Microsoft Exchange Integration is enabled and the phone is connecting to Exchange Server 2010 or later, Exchange Web Services (EWS) is used for the connection. This connection is secured using HTTPS by default which means that the phone is required to validate the Exchange Server identity certificate. To validate the certificate, the TRUSTCERTS parameter in the settings file must include the root certificate of the Certificate Authority (CA) which issued the Exchange Server identity certificate. This configuration will work if the identity certificate was directly issued by the CA root certificate.

If a public CA such as VeriSign is used to obtain an identity certificate for the Exchange Server, the identity certificate will be issued by an intermediate CA certificate and not by the root. In this case, both the root and intermediate CA certificates must be installed on the phone using TRUSTCERTS or the HTTPS connection will fail. In general, if the Exchange Server identity certificate is issued by an intermediate CA, all certificates from the intermediate CA up to the

root must be included in TRUSTCERTS for installation on the phone so that the entire certificate chain is available for validation.

Debug mode

As a general guide, it should be noted that response times could be impacted when debug or syslog is enabled

Do not enable debug level logs unless you are debugging an issue, phone performance will be noticeably slow if too many debug categories are enabled.

SIP CONTROLLER LIST

This parameter consolidates SIP controller parameters for IP address, port, and transport protocol into a single configuration parameter. The parameter setting should be a list of controller information where the format for each controller entry is "host:port;transport=xxx". The host should be specified only by an IP address when interworking with Avaya Aura™. This applies to all sources of the SIP_CONTROLLER_LIST parameter which includes DHCP, LLDP, Web interface and the 46xxsettings.txt file.

Security Certificates – IP Address versus FQDN

There is an industry movement towards the use of a FQDN (Fully Qualified Domain Name) instead of an IP address for the Subject Alternate Name or Subject Common Name for security certificates. J100 software supports a FQDN_IP_MAP parameter which specifies mapping of FQDNs to IP addresses for the purpose of validating an FQDN identity found in a server certificate.

Note: Starting in 4.0.11, the phone supports use of FQDN when connecting to Aura so the use of FQDN_IP_MAP to get proper validation of server certificate identity is no longer required.

SRTP (Media Encryption)

In order to correctly use SRTP, there are various components within the network that you must correctly configure. For J100 Series IP Phones to function properly with SRTP in an Avaya Aura© environment, you must configure the equivalent parameters in Communication Manager or System Manager. Avaya strongly recommends that the following three parameters on the J100 Series IP Phones and the equivalent Communication Manager parameters must match:

SET ENFORCE_SIPS_URI 1
SET SDPCAPNEG 1
SET MEDIAENCRYPTION X or
SET MEDIAENCRYPTION X,Y or
SET MEDIAENCRYPTION X,Y,Z

J100 software supports AES-256 media encryption. Care must be taken to properly configure the encryption parameter when this is used in conjunction with other devices that do not support AES-256.

Multi Device Access

Refer to the <u>"Avaya Aura Multi Device Access White Paper"</u> which is available on http://support.avaya.com for known limitations.

Language support

The J129 IP Phones does not support an Arabic, Thai, or Traditional Chinese user interface.

Language localization tool and additional language packages can be found at support.avaya.com site <u>here</u>.

Ringtone and Ringtone Wave Files

Numeric only naming conventions should be avoided with ringtone names (E.g. 12345.wav). The maximum allowed size of an individual ringtone file is 512 KB. The maximum allowed size of all ringtone files is 5120 KB.

Headset Profiles

J100 SIP software supports "Headset Profiles" to provide optimum performance for different brands of headsets. An up-to-date version of the profile <-> vendor cross reference can be found at https://downloads.avaya.com/css/P8/documents/100173755.

Avaya Session Border Controller for Enterprise

For all IP Phones which are remotely connected through an SBCE, please ensure that the following is set in the 46xxsettings.txt file

SET WAIT FOR REGISTRATION TIMER 40

SIP Transport Protocols

TCP or TLS are the recommended transport protocols. UDP transport is not supported with J100 SIP software except in a OpenSIP environment.

Encryption – SHA2 and RSA 2048

J100 software supports RSA 2048-bit length encryption keys and supports the SHA2 (224, 256, 384, and 512) hash algorithms. This has been certified for HTTPS usage for web-based administration of these phone sets. When the TLS server-client handshake is initiated, this IP Phone (operating as the client) is able to send its Identity certificate with an enhanced digital signature (SHA2/2048 key). Additionally, this IP Phone is able to receive and validate server Identity certificates which have an enhanced digital signature (SHA2/2048 key).

Interworking – Avaya Diagnostic Server (ADS)

¹ J129 does not support a headset

Avaya J100 SIP Release 2.0.0.0 and later supports the ADS server. The SLMSRVR parameter must be set in the 46xxsettings.txt file for this version of the agent to register with ADS. In addition, a valid certificate file must be downloaded via TRUSTCERTS.

Avaya Diagnostic Server 3.0.3 is the minimum release to support the J129 IP Phone and the J179 IP Phone.

Avaya Diagnostic Server 3.0.4 is the minimum release to support the J139 IP Phone.

Avaya Diagnostic Server 3.1.0 is the minimum release to support the J159 IP Phone.

Avaya Diagnostic Server 3.1.1 is the minimum release to support the J189 IP Phone.

Avaya Diagnostic Server 3.2 is the minimum release to support remote worker for the J159 IP Phone and the J189 IP Phone.

"Desk Phone" Mode and Lock

Avaya one-X® Communicator, Avaya Workplace and similar UC applications from Avaya support a "Desk Phone" (Shared Control) mode in which the UC application can control an associated IP Phone. An IP Phone supports a "Lock" mode, which can be entered either manually or automatically, which prevents the dialing of any number except for an emergency number using the keypad of the IP Deskphone. If an IP Phone is in Shared Control with a UC application and is also in a "Lock" state, placing a call from the UC application will still result in the call being established from the IP Phone.

Interworking – TLS 1.3

Starting with release 4.1.1.0, J100 software supports TLS 1.3.



J100 software also includes a configuration parameter (TLS_VERSION) which can be used to configure the IP Phone to <u>only</u> use TLS 1.3. Care must be taken to only use this parameter when all components to which the IP Phone will communicate using TLS also support TLS 1.3.

J129 - Presence

The J129 does not display presence in an Avaya Aura® network or have the ability to manually set a presence state. The J129 publishes presence information for other clients that support viewing presence.

The J139, J159, J179 and J189 display presence, publish presence, and can manually set a presence state.

VLAN separation

The J100 software supports 3 versions of VLAN separation; 1) Full VLAN separation, 2) Partial VLAN separation and 3) No VLAN separation. However, the J129 IP Phone does NOT support partial VLAN separation.

Avaya highly recommends that voice and data traffic be separated by VLANs and that voice traffic has its own VLAN.

Features not supported on the J129 Phone

The following features are not supported by the J129 IP Phone with J100 software:

- Exchange integration, WML browser, URI dialing, simultaneous display of caller name and number, redial by list, conference roster list, missed call filtering, displaying presence, downloadable ringtones, Favorites, Personalize labels
- Bridge call appearances (except MDA)
- MLPP, Call Pickup, Hunt Group Busy, Team Button, Enhanced Call Forward, Dial Intercom, Exclusion, LNCC, Priority Calls, Whisper Page, Busy Indicator
- Interworking with Contact Center Elite (CC Elite)
- Bluetooth

Features not supported on the J139 Phone

The following features are not supported by the J139 IP Phone with J100 software:

- Interworking with Contact Center Elite (CC Elite)
- Wifi and Bluetooth

Features not supported on the J159 Phone

The following features are not supported by the J159 IP Phone with J100 software:

• Interworking with most Contact Center Elite (CC Elite) (Agent login and logout is supported)

J129 with IP Office – Features supported

The following features are supported by the Avaya J129 IP Phone when deployed on Avaya IP Office™: Attended transfer, Unattended transfer, transferring a call by selecting a contact or recents, personal directory, voice mail, manual dial mode, conference.

The following features are supported by the Avaya J129 IP Phone when deployed on Avaya IP Office™ using a **short code**: Call Forward, Call Forward Busy, Call Park/Unpark, Do Not Disturb, Automatic Call Back, Private Call, Speed Dial.

J139 with IP Office – Features supported / not supported

The following features are supported by the Avaya J139 IP Phone when deployed on IP Office TM :

- Basic call handling on Call Appearances and Line Appearances only Making a call,
 Call presentation, Answer, Hold, Transfer, Conference, Drop
- IP Office Directory (Personal and System)

- IP Office Call History
- Visual Voice

Include basic operation and call handling feature controls by default via IP Office Features Menu

- DND
- Forwarding
- Mobile Phone Call Twinning (User must first be administered to permit Mobile Twinning by a system Administrator).
- Hot Desking

Allow basic call handling feature controls to be administered as button features by a system Administrator

- Call Park
- Call Pickup
- Call Page
- Call Recording
- Auto Call-back
- Account Code
- Authorization Code
- User BLF (NOTE: Requires IP Office 11.0 FP4)
- Group BLF (*NOTE: Requires IP Office 11.0 FP4*)

Allow basic agent controls to be administered as button features by a system Administrator

- Hunt Group Membership
- Agent Status
- After Call Work
- Coaching Request

The following features are <u>not</u> supported by the Avaya J139 IP Phone when deployed on IP Office™:

Advanced Call Presentation / Handling:

- MADN
- Bridged Appearances
- Coverage Appearances

IP Office Features/Status Menus:

- Advanced Call Pickup
- Advanced Call Park
- DND exceptions
- Account / Authorization Code
- Auto Answer Controls
- Withhold Number
- Coverage Ring Controls
- Advanced Hunt Group Controls: (Multi Membership, Group Status, Group Configuration)
- Self-Administration
- System Administration

Button configuration:

- Hands-free Answer
- Automatic Intercom
- Specific Call Dial Types
- Conference Meet-Me
- Self-Administration
- System Administration
- Advanced Hunt Group Controls (Group Status, Group Configuration)
- Agent Supervisor Features: (Call Steal, Call Listen, Call Intrude, Coaching Intrusion)

Others:

- 9600/J100 Push API
- WML Browser
- Exchange Calendar/Contact Integration
- Multicast Paging

J159/J179/J189 with IP Office - Features supported / not supported

The following features are supported by the Avaya J159/J179 IP Phone when deployed on IP Office™:

- Basic call handling on Call Appearances and Line Appearances only Making a call,
 Call presentation, Answer, Hold, Transfer, Conference, Drop
- IP Office Directory (Personal and System)
- IP Office Call History
- Visual Voice

Include basic operation and call handling feature controls by default via IP Office Features Menu

- DND
- Forwarding
- Mobile Phone Call Twinning (User must first be administered to permit Mobile Twinning by a system Administrator).
- Hot Desking

Allow basic call handling feature controls to be administered as button features by a system Administrator

- Call Park
- Call Pickup
- Call Page
- Call Recording
- Auto Call-back
- Account Code
- Authorization Code
- User BLF
- Group BLF

Allow basic agent controls to be administered as button features by a system Administrator

- Hunt Group Membership
- Agent Status
- After Call Work
- Coaching Request

The following features are also supported by the Avaya J159/J179 IP Phone when deployed on IP Office™:

Advanced Call Presentation / Handling:

- MADN
- Bridged Appearances
- Coverage Appearances

IP Office Features/Status Menus:

- Advanced Call Pickup
- Advanced Call Park
- DND exceptions
- Account / Authorization Code
- Auto Answer Controls
- Withhold Number
- Coverage Ring Controls
- Advanced Hunt Group Controls: (Multi Membership, Group Status, Group Configuration)
- Self-Administration
- System Administration

Button configuration:

- Hands-free Answer
- Automatic Intercom
- Specific Call Dial Types
- Conference Meet-Me
- Self-Administration
- System Administration
- Advanced Hunt Group Controls (Group Status, Group Configuration)
- Agent Supervisor Features: (Call Steal, Call Listen, Call Intrude, Coaching Intrusion)

The following features are <u>not</u> supported by the Avaya J159/J179 IP Phone when deployed on IP Office $^{\text{TM}}$:

- Personalization (i.e. ability to reconfigure the button layout)
- 9600/J100 Push API
- WML Browser
- Exchange Calendar/Contact Integration
- Multicast Paging

Deploying the J129/J139/J159/J179/J189 in OpenSIP Platform

The J129/J139/J159/J179/J189 are supported with Broadsoft Broadworks, Asterisk, FreeSwitch, and Netsapiens. IP phone configuration file (settings file) must be deployed

from a file server (HTTP or HTTPS). User backup/restore must also be deployed from a file server (HTTP or HTTPS). SIP Transport = TLS is not supported. For these phones to work in an OpenSIP environment, configuration file (settings file) must have following parameter configured with value as given:

- SET ENABLE_AVAYA_ENVIRONMENT 0
- SET DISCOVER_AVAYA_ENVIRONMENT 0
- SET ENABLE IPOFFICE 0

See "Installing and Administering Avaya J100 Series IP Phones in an Open SIP environment" for more detail.

Provisioning of File Server Address

Phone can be provisioned using HTTP/S File Server. The HTTP/S File Server address can be provided to the phone through one of the following methods:

- DHCP
- LLDP
- CRAFT/Web Interface
- Device Enrolment Service (DES)

HTTPS file server has priority over the HTTP file server if both configured.

Once provisioned using one of the above methods, HTTP/S file server address can also be changed through settings file by using following parameters:

- For HTTP → HTTPSRVR, HTTPDIR, HTTPPORT
- For HTTPS → TLSSRVR, TLSDIR, TLSPORT

Once File server address is changed through settings file it will override the file server address provided through DHCP or LLDP. Thus, it is advised to use this option only if different server address needs to be provided to override the DHCP.

If HTTPS file server address is configured in setting file, phone will contact to HTTPS server immediately after the download of settings file without any reboot.

Note:

Please take a note that when HTTPS file server address is configured in settings file, configure SET HTTPSRVR "" in the settings file to override the HTTPSRVR value received from DHCP. Commenting out the HTTPSRVR parameter will not override the value received from DHCP.

Preconfig keys considerations

Added in Release 4.0.2.0 and improved in Release 4.0.8.0 is the ability to pre-configure keys using the 46xxsettings.txt file or the WebUI.

Due to multiple ways you can configure keys in an Aura environment please be aware if you are configuring keys in System Manager, WebUI, and the 46xxsettings.txt file that conflicts may arise.



Note: If you downgrade your phone software from version 4.0.8.0 or greater to a version less than 4.0.8.0, users lose their labels and favorites, modified on the phone and administrators lose labels and favorites, modified through the phone web interface.

Note: MDA groups must use the same preconfig.

For more details please refer to the "Installing and Administering Avaya J100 IP Phones".

PHONEKEY customization limitations

CCElite phones are not supported by PHONEKEY customization.

Recents (Call History) – limitations when downgrade from 4.0.3.0 or later

Starting in J100 SIP 4.0.3.0, call logs are encrypted on the phone. If the phone is downgraded from 4.0.3.0 to any previous release, then the **call logs will be lost** since the older release will not be able to read the encrypted logs. This will occur in all environments except Avaya Aura® when centralized call logging is enabled. If centralized call logging is enabled, then Recents will be preserved on downgrade from 4.0.3.0 to older firmware versions.

Exchange Calendar Integration migrating to OAuth authentication

Exchange Integration basic authentication for EWS is not supported for newly created tenants as of Oct 13, 2020. This is replaced by OAuth authentication which is supported on J100 in Release 4.0.7.0 and greater. Please see the user guide for further details.

USB Headset support

J100 J189 and J159 provide support for standard USB HID headsets. This includes basic call control features.

Depending on the Headset vendor implementation, model specific features may or may not work. Support of feature set may be limited by specific headset model capabilities.

Upgrade to SIP 4.0.10 and greater from SIP 4.0.8.0

J100 SIP phones being upgraded to 4.0.10 and greater from 4.0.8.0 must upgrade to 4.0.9.0 first.

NOTE: Although direct upgrade from 4.0.8 to 4.0.10 and greater may work for phones fresh out of the box, it may fail for some existing phones that are in use.

SSO considerations

Lock screen Application

If the lock screen application is enabled (ENABLE_PHONE_LOCK = 1) or lock screen timer (PHONE_LOCK_IDLETIME > 0) are enabled, the user will need to know the

password to unlock the phone. Although this is no different than without SSO the end user is less likely to know the password since they did not use it to login. If you enable locking then make sure to enable the user lock PIN by setting LOCK_MODE to 1, so the end user can define and manage their lock pin.

Setting GROUP from System Manager

If users will be regularly logging out and logging back in then it is not recommended to assign GROUP to the phone in System Manager. Doing this will cause the phone to reboot after SSO login causing a longer login process.

802.1x EAP TLS changes

Prior to R4.1.1 the J100 SIP phones will always start the 802.1x supplicant when EAP is enabled.

Starting in R4.1.1 the J100 SIP phones will not start the 802.1x supplicant if there is no identity certificate or any trust certificate installed on the phone and EAP is enabled with method TLS.

See DOT1XEAPTLSONLYWITHCERT parameter for further details.

J100 4.1.9.0 Resolved Issues (since J100 4.1.8.0)

The following table includes issues which are resolved with this release of software compared to ${\tt J100~4.1.8.0.8}$

External ID	Internal ID	Issue Description		
Avaya Aura®				
1-22995759762	SIP96X1-112863	SMGR 10.1.3.5 J179 favorite button disappears		
1-AKOVHVE				
1-22896335132	SIP96X1-112344	J100 station plays a tone to the user, new inbound		
1-AJGHB2H		calls to the station do not stop the tone.		
1-22923586831	SIP96X1-112174	Mass users reporting button labels issue missing		
1-AJ440VM		after network outage of flap		
1-22904283732	SIP96X1-112120	Europe timezone has wrong data. DST started 1		
1-AIUDSE2	A	week early in Europe from 23rd Sunday		
	Avay	a Cloud Office™		
		IP Office™		
		CCElite		
1-AGH3EKT	SIP96X1-111808	CALL TRANSFER: DISPLAY: Originator/Customer		
		Phone Number is not displayed when call is		
		transferred to another Agent.		
	Open SIP			
	A	II Platforms		
	Wifi			
Web User Interface				
	Button	Module (JEM24)		

Unresolved issues in J100 4.1.9.0

The following table includes unresolved issues with this release of software which were known as of the issue date of this document.

External ID	Internal ID	Issue Description
		Avaya Aura [®]
	SIP96X1-112192	The phone keeps Primary lock state after Guest SSO
		user login
		work-around: unlock with Guest credentials
	•	Avaya Cloud Office™
	SIP96X1-110536	A J139 phone is not registered back to ACO server after
		failover to IPO and failback to ACO. IP Office™
	Γ	11 011100
		Open SIP
	SIP96X1-41164	HTTP redirect to HTTPS fails certificate validation,
		connection fails
		work-around: reconfigure phone URL to HTTPS
	SIP96X1-66640	Do not to use Web UI>Environment Settings>3PCC
		Server Mode = Netsapiens even though it shows, use
		generic All Platforms
	GTD06\/4 00004	
	SIP96X1-89301	The phone does not download TRUSTCERTS list when
		user defines a HTTPs URL to the phone (SET ENABLE_PUBLIC_CA_CERTS=1 in settings file)
		work-around: use HTTP or SET
		ENABLE_PUBLIC_CA_CERTS=0
	SIP96X1-88631	L149 USB mute control drops call
		work-around: use mute control on phone, this is a
		wired USB headset
	SIP96X1-113194	Phone displays duplicated Languages in Menu setting if
		lang file is customized Web User Interface
	T	
	SIP96X1-94382	Language Dutch imported from WebServer is not
		displayed on phone UI
	SIP96X1-102120	work-around: Import Dutch language file from settings file Customized softkey is not displayed if it was earlier set
	31F90X1-102120	from Web UI and it is removed before setting it via
		46xxsettings on J100 phone.
		work-around: Page level reset to default needs to be
		done to completely get rid of the softkey configured
		from WEB
	T	Bluetooth
	SIP96X1-105473	No incoming call alert and cannot answer the call via
	D.	Bluetooth when Incoming call popup is disabled. utton Module (JEM24)
	DI	utton riodule (JEM24)

External ID	Internal ID	Issue Description
Wifi		

Appendix 1 – Supported Hardware and Minimum Software Release

J100 SIP 4.1.9.0 software is supported on the following models of IP Phones. Models may ship from the factory with a different load of software pre-installed. As such, they should be upgraded to this release on first installation.

Note: Comcodes indicated with an asterisk (*) have an End-of-Sale Notification and include a link to the corresponding end-of-sale document.

Comcode	Short Description	Model(s)	Replaced by
700512392 *	J129 IP PHONE	J129D01A	700513638 700517252
700513638 *	J129 IP PHONE NO PWR SUPP	J129D02A	700517252 700514813 700517252
700512969 *	J129 IP PHONE 3PCC W/O PWR SUPP	J129D01A	700513639 700517252
700513639 *	J129 IP PHONE 3PCC W/CERT	J129D02A	700514814 700517252
700514813 *	J129 IP PHONE 5V	J129D03A	700513916 700517252
700515186 *	J129 IP PHONE 5V ENCRYPTION DISABLED	J129D03X	700515187
700514814 *	J129 IP PHONE 5V 3PCC	J129D03A	700513917 700517252
700517252	J129 IP PHONE GLOBAL	J129D03A	
700513916	J139 IP PHONE GLOBAL	J139D01A J139D01B	
700515187	J139 IP PHONE ENCRYPTION DISABLED	J139D01X	
700513917 *	J139 IP PHONE 3PCC	J139D01A J139D01B	700513916
700513918	J139 IP PHONE TAA	J139D01A J139D01B	
700514634	J139 IP PHONE ORANGE	J139D01A J139D01B	
700512394	J159 IP PHONE GLOBAL	J159D01A J159D01B	
700515188	J159 IP PHONE ENCRYPTION DISABLED	J159D01X	
700512395	J159 IP PHONE TAA	J159D01A J159D01B	
700512970 *	J159 IP PHONE 3PCC	J159D01A	700512394
700515582	J159 IP PHONE ORANGE	J159D01A J159D01B	
700513634 *	J169 IP PHONE NO PWR SUPP	J169D01A J169D01B J169D02A	700513569
700515189 *	J169 IP PHONE ENCRYPTION DISABLED	J169D01X J169D02X	700515190
700513635 *	J169 IP PHONE TAA	J169D01A J169D01B	700513629

Comcode	Short Description	Model(s)	Replaced by
		J169D02A	
700513636 *	J169 IP PHONE 3PCC	J169D01A	700513630
		J169D01B	
		J169D02A	
700514468 *	J169 IP PHONE GLOBLE WHITE	J169D01A	700514469
		J169D01B	
		J169D02A	
700514757 *	J169 IP PHONE NO BEZEL	J169D01A	700514758
		J169D01B	
		J169D02A	
700514635 *	J169 IP PHONE ORANGE	J169D01A	
		J169D01B	
		J169D02A	
700513569	J179 IP PHONE GLOBAL	J179D02A	
		J179D03A	
		J179D03B	
700515190	J179 IP PHONE ENCRYPTION DISABLED	J179D02X	
		J179D03X	
700513629	J179 IP PHONE TAA	J179D02A	
		J179D03A	
		J179D03B	
700513630 *	J179 IP PHONE 3PCC	J179D02A	700513569
		J179D03A	
		J179D03B	
700514469	J179 IP PHONE GLOBLE WHITE	J179D02A	
		J179D03A	
		J179D03B	
700514636	J179 IP PHONE ORANGE	J179D02A	
		J179D03A	
		J179D03B	
700514758	J179 IP PHONE NO BEZEL	J179D02A	
		J179D03A	
		J179D03B	
700512396	J189 IP PHONE GLOBAL	J189D01A	
		J189D01B	
700512397	J189 IP PHONE TAA	J189D01A	
		J189D01B	
700512971 *	J189 IP PHONE 3PCC	J189D01A	700512396
700515191	J189 IP PHONE ENCRYPTION DISABLED	J189D01X	

Avaya periodically releases new hardware variations of the J100-Series IP Phones typically to address a need to change hardware components. That change may require a new version of software to support the new hardware. This then forces a minimum software release supported on that hardware. **Attempts to downgrade these models to lower versions of software will be rejected**.

The following table provides a matrix of the different models of J100 Series IP Phones including hardware generations and any limitation on supported software version. The "Model" information can be found on the label on the outside of the shipping box, on the label on the back of the IP Phone, within the Information menus available from the screen of the phone, remotely via LLDP, remotely via the Web Interface (SIP software), and remotely via SNMP. The "Hardware Revision" can be found on the label of the box on the same line as the "Model" information, and also found on the label on the back of the phone on the first row.

Model	Hardware	Minimum SIP	Minimum H.323
	Revision(s)	Software	Software
J129D01A	all	1.0.0.0.43	N/A
J129D02A	01 to 07	2.0.0.0.45	N/A
J129D02A	08 and greater	3.0.0.1.6	N/A
J129D03A	01 to 11	4.0.2.0.8	N/A
J129D03A	12 and greater	4.0.3.1.4	N/A
J139D01A	01 to 05, 07, 08	3.0.0.0.20	N/A
J139D01A	06, 09 to 11	3.0.0.1.6	N/A
J139D01A	12 to 20	4.0.0.0.21	N/A
J139D01A	21 to 24	4.0.3.1.4	N/A
J139D01A	25 to 30	4.0.8.0.13	N/A
J139D01A	31 to 33	4.0.11.0.3	N/A
J139D01B	34 and greater	4.0.12.0.6	N/A
J159D01A	01 to14	4.0.3.1.4	6.8.5.02
J159D01B	15 and greater	4.0.12.0.6	6.8.5.3.2
J169D01A	01 to 03	1.5.0.0.15	6.7.0.02
J169D01A	04 to 07	3.0.0.1.6	6.8.0.03
J169D01B	08 to 18	4.0.0.0.21	6.8.0.03
J169D01B	19 and greater	4.0.3.1.4	6.8.3.04
J179D02A	01 to 03	1.5.0.0.15	6.7.0.02
J179D02A	04 to 08	2.0.0.0.45	6.7.0.02
J179D03A	09 to 11	4.0.1.0.11	6.8.2.02
J179D03A	12 to 18	4.0.2.0.8	6.8.2.02
J179D03A	19 to 33	4.0.3.1.4	6.8.3.04
J179D03B	34 and greater	4.0.12.0.6	6.8.5.3.2
J189D01A	01 to 15	4.0.6.1.4	6.8.5.02
J189D01B	16 and greater	4.0.12.1.x	6.8.5.3.2

Appendix 2 - Release History

The following table provides a history of the J100 SIP software releases. The "ID" column shows the identifier of this software which is seen in the "About" menu item.

Release	ID	Date	Link to Readme file
1.0.0.0	1.0.0.0.43	Dec 2016	https://support.avaya.com/css/P8/documents/101033485
1.1.0.0	1.0.0.0.15	Mar 2017	https://support.avaya.com/css/P8/documents/101037079
1.1.0.1	1.0.0.1.3	Aug 2017	https://support.avaya.com/css/P8/documents/101042514
1.5.0.0	1.5.0.0.15	Mar 2018	https://support.avaya.com/css/P8/documents/101047039
2.0.0.0	2.0.0.0.45	April 2018	https://support.avaya.com/css/P8/documents/101048016
3.0.0.0	3.0.0.0.20	July 2018	https://support.avaya.com/css/P8/documents/101050223
3.0.0.1	3.0.0.1.6	Aug 2018	https://support.avaya.com/css/P8/documents/101051793
3.0.0.2	3.0.0.2.2	Nov 2018	https://support.avaya.com/css/P8/documents/101053115
4.0.0.0	4.0.0.0.21	Dec 2018	https://support.avaya.com/css/P8/documents/101054005
4.0.0.1	4.0.0.1.2	Mar 2019	https://support.avaya.com/css/P8/documents/101056162
4.0.1.0	4.0.1.0.11	Apr 2019	https://support.avaya.com/css/P8/documents/101056525
4.0.2.0	4.0.2.0.8	July 2019	https://support.avaya.com/css/P8/documents/101058668
4.0.2.1	4.0.2.1.3	July 2019	https://support.avaya.com/css/P8/documents/101059981
4.0.3.0	4.0.3.0.10	Oct 2019	https://support.avaya.com/css/P8/documents/101060975
4.0.3.1	4.0.3.1.4	Nov 2019	https://support.avaya.com/css/P8/documents/101062454
4.0.4.0	4.0.4.0.10	Jan 2020	https://support.avaya.com/css/P8/documents/101063151
4.0.5.0	4.0.5.0.10	Apr 2020	https://support.avaya.com/css/P8/documents/101065323
4.0.6.0	4.0.6.0.7	June 2020	https://support.avaya.com/css/P8/documents/101068496
4.0.6.1	4.0.6.1.4	Aug 2020	https://support.avaya.com/css/P8/documents/101070109
4.0.6.1	4.0.6.1.6	Sep 2020	https://support.avaya.com/css/P8/documents/101070565
4.0.7.0	4.0.7.0.7	Oct 2020	https://support.avaya.com/css/P8/documents/101071218
4.0.7.1	4.0.7.1.5	Dec 2020	https://support.avaya.com/css/P8/documents/101072194
4.0.8.0	4.0.8.0.14	Mar 2021	https://support.avaya.com/css/P8/documents/101074480
4.0.9.0	4.0.9.0.4	Apr 2021	https://support.avaya.com/css/P8/documents/101075292
4.0.10.0	4.0.10.0.4	July 2021	https://support.avaya.com/css/P8/documents/101076605
4.0.10.1	4.0.10.1.2	Aug 2021	https://support.avaya.com/css/P8/documents/101077135
4.0.10.2	4.0.10.2.1	Oct 2021	https://support.avaya.com/css/P8/documents/101078268
4.0.10.2	4.0.10.2.2	Dec 2021	https://support.avaya.com/css/P8/documents/101078268
4.0.10.3	4.0.10.3.2	Dec 2021	https://support.avaya.com/css/P8/documents/101079719
4.0.11.0	4.0.11.0.3	Feb 2022	https://support.avaya.com/css/P8/documents/101080577
4.0.12.0	4.0.12.0.6	Apr 2022	https://support.avaya.com/css/P8/documents/101081376
4.0.12.1	4.0.12.1.1	May 2022	https://support.avaya.com/css/P8/documents/101081837
4.0.13.0	4.0.13.0.6	July 2022	https://support.avaya.com/css/P8/documents/101082968
4.0.14.0	4.0.14.0.7	Oct 2022	https://support.avaya.com/css/P8/documents/101083795
4.1.0.0	4.1.0.0.7	Jan 2022	https://support.avaya.com/css/P8/documents/101084676

Release	ID	Date	Link to Readme file
4.1.1.0	4.1.1.0.7	Apr 2022	https://downloads.avaya.com/css/P8/documents/101085556
4.1.2.0	4.1.2.0.11	Oct 2023	https://downloads.avaya.com/css/P8/documents/101087001
4.1.2.1	4.1.2.1.1	Feb 2024	https://downloads.avaya.com/css/P8/documents/101087449
4.1.3.0	4.1.3.0.6	Jan 2024	https://downloads.avaya.com/css/P8/documents/101088004
4.1.4.0	4.1.4.0.5	Apr 2024	https://downloads.avaya.com/css/P8/documents/101088985
4.1.5.0	4.1.5.0.6	July 2024	https://downloads.avaya.com/css/P8/documents/101090998
4.1.6.0	4.1.6.0.6	Oct 2024	https://downloads.avaya.com/css/P8/documents/101091653
4.1.7.0	4.1.7.0.5	Jan 2025	https://downloads.avaya.com/css/P8/documents/101092321
4.1.8.0	4.1.8.0.8	Apr 2025	https://downloads.avaya.com/css/P8/documents/101092697

Appendix 3 - New and changed 46xxsettings.txt parameters

The latest version of the 46xxsettings.txt file can be downloaded from

https://support.avaya.com/downloads/download-details.action?contentId=C201773928555860 8&productId=P1661

New parameters

```
####### GUEST LOGIN (AND VISITING USER) SETTINGS ########
## ALLOW_GUEST_SESSION_PERSISTENCE specifies whether Guest session persistence will be allowed after reboot.
## Value Operation
## 0 Disabled (default)
## 1 Enabled
## This parameter is supported by:
## J139, J159, J169, J179, J189 SIP R4.1.9.0 and later only in Aura Environment
## SET ALLOW_GUEST_SESSION_PERSISTENCE 1
## ENABLE_MD5 Specifies whether the phone supports the MD5 algorithm for SIP Digest Authentication.
## The phone will always support additional digest authentication methods as defined by RFC 8760, including:
## SHA-256
## SHA-512/256
## Before disabling this parameter, ensure that your server is configured to request at least 1 of the above methods
## Note: The phone will select the first supported digest authentication method from the list of methods provided by the server.
## Value Operation
## 0 Disabled
## 1 Enabled (default)
## This parameter is supported by:
## I100 SIP R4.1.9.0 and later
## SET ENABLE_MD5 0
##
## PREFERRED_CONTACTURI_NUMBER Specifies to take number from contact header during transfer.
## Value Operation
## 0 Disabled (default)
## 1 Enabled
## This parameter is supported by:
## J100 SIP R4.1.9.0 and later
## SET PREFERRED_CONTACTURI_NUMBER 1
## CALL_RECORDING_ENABLED Specifies if the phone will provide the user a Call Recording feature button. Allows the user to activate,
pause and resume call recording of an active call session.
## This parameter only applies to Avaya Cloud Office server environment.
## Value Operation
## 0 Disabled (Default)
## 1 Enabled
## This parameter is supported by:
## [139, [159, [169, [179, [189 SIP R4.1.9.0 and later, only in ACO environment
## SET CALL_RECORDING_ENABLED 1
```

Changed parameters

```
## If the feature is set to enabled. phone will keep the current line selection. If the feature is set to disabled, phone will select a higher
priority call appearance
## or reset to select the first line (if the phone is idle.). The call can be on a primary call appearance, a bridged call appearance or a
shared call appearance.
## When the feature is enabled or disabled without 'Forced' values (0 or 1), the Phone user is permitted to override this value. If the
feature is enabled or disabled
## using 'Forced' values (2 or 3), the Phone user is not allowed to change this value.
## Value Operation
## 0 Disable - select a higher priority call or reset to select the first line if the phone is idle
## 1 Enable - keep the current line selection (default)
## 2 Forced Disable
## 3 Forced Enable
## This parameter is supported by:
     [100 SIP R4.0.4.0 and later ([139, [159, [169, [179 only), [189 SIP R4.0.6.1 and later values 0 and 1 are supported.
      J100 SIP R4.1.9.0 and later (J139, J159, J169, J179, J189 only) along with 0 and 1, values 2 and 3 are also supported.
## SET KEEP_CURRENT_CA 0
##
## SOFTKEY_TRANSFER_OUTGOING specifies new custom softkey for CA lines in TRANSFER OUTGOING state.
## The default value is "
## "Blank" and "Function" are supported as type. When "Function" is used, then the action is: "complete", "cancel" and "autocbck".
## Note: Field names ("type", "action", "label") are case insensitive (for example, "action", "ACTION", "Action" are all supported). "type"
value - always case insensitive.
## "action" value - case insensitive for "function" type actions, case sensitive for "dial" type actions. "label" values - always case
sensitive.
## Note: "blank" softkeys do not require action or label fields. "Blank" is an empty softkey with no function, which can be used to fill in a
location in a softkey set. "Blank" automatically replaces softkeys
## which are not allowed by either configuration or runtime conditions.
## Note: When using "ADD SOFTKEY_TRANSFER_OUTGOING" the order of the lines will be used for presentation of softkeys for CA lines
in TRANSFER OUTGOING state.
## Note: There is an option to use one single line of "SET" with multiple softkeys. e.g. "SET SOFTKEY_TRANSFER_OUTGOING
"type=function;action=complete;label=complete,type=function;action=cancel;label=cancel"
## Note: SET keyword overrides the value, while ADD adds content into the end of previous value separated by comma. The first record
can be both SET and ADD, but following records must use ADD keyword.
## Note: For the complete list of type/action, please refer to the administration guide.
## This parameter is supported by:
      J100 SIP R4.0.7.1 and later (Not supported by J129) (supported in OpenSIP and Avaya Aura environments (except Avaya IP
##
Office))
      [100 SIP R4.1.9.0 and later (Not supported by [129, supported in OpenSIP and Avaya Aura environments (except Avaya IP Office)],
support is added to set AutoCbck custom softkey for outgoing transfer
## ADD SOFTKEY_TRANSFER_OUTGOING "type=function; action=complete; label=Complete"
## ADD SOFTKEY_TRANSFER_OUTGOING "type=function;action=cancel;label=Cancel"
## ADD SOFTKEY_TRANSFER_OUTGOING "type=function;action=autocbck;label=AutoCbck"
## ADD SOFTKEY_TRANSFER_OUTGOING "type=dial;action=123456;label=Dial1"
##
## SOFTKEY_CONFERENCE_OUTGOING specifies new custom softkey for CA lines in CONFERENCE OUTGOING state.
## The default value is "".
## "Blank" and "Function" are supported as type. When "Function" is used, then the action is: "cancel" and "autocbck".
## Note: Field names ("type", "action", "label") are case insensitive (for example, "action", "ACTION", "Action" are all supported). "type"
value - always case insensitive.
## "action" value - case insensitive for "function" type actions. "label" values - always case sensitive.
## Note: "blank" softkeys do not require action or label fields. "Blank" is an empty softkey with no function, which can be used to fill in a
location in a softkey set. "Blank" automatically replaces softkeys
## which are not allowed by either configuration or runtime conditions.
## Note: When using "ADD SOFTKEY_CONFERENCE_OUTGOING" the order of the lines will be used for presentation of softkeys for CA
lines in CONFERENCE OUTGOING state.
## Note: SET keyword overrides the value, while ADD adds content into the end of previous value separated by comma. The first record
can be both SET and ADD, but following records must use ADD keyword.
## Note: For the complete list of type/action, please refer to the administration guide.
## This parameter is supported by:
      J100 SIP R4.0.7.1 and later (Not supported by J129) (supported in OpenSIP and Avaya Aura environments (except Avaya IP
##
Office))
      J100 SIP R4.1.9.0 and later (Not supported by J129, supported in OpenSIP and Avaya Aura environments (except Avaya IP Office)),
support is added to set AutoCbck custom softkey for outgoing conference
## ADD SOFTKEY_CONFERENCE_OUTGOING "type=function;action=cancel;label=Cancel"
## ADD SOFTKEY_CONFERENCE_OUTGOING "type=function;action=autocbck;label=AutoCbck"
```

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