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SIP Server Deployment Guide

Supplement to SIP Server Deployment Guide

5/4/2025

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This supplement provides descriptions of new features introduced in SIP Server 8.1.1 as part of the SIP Server Continuous Delivery project. For information on new or updated configuration options, see [here](#).

The current documentation set for SIP Server can be found [here](#).

8.1.1 Features Support

The following features are described in this supplement:

Released in Version	Feature Name/Article	Date Released
8.1.104.49	Re-establishing connection on TCP transport exception	June 28, 2022
8.1.104.34	Remote agent deletion from a conference (TDeleteFromConference)	November 23, 2021
8.1.103.95	HTTP Monitoring Interface is updated to include statistics for T-Library client connections of the following SIP Server threads statistics : Session Controller (the main T-Server thread), T-Controller, Interaction Proxy, and Smart Proxy	July 21, 2020
8.1.103.88	Masking/unmasking specific SIP headers contained in SIP Server logs	May 12, 2020
	Enabling office-based agents to work from home - sample configuration	March 27, 2020
8.1.103.80	Enhanced handling of XS requests	March 20, 2020
8.1.103.35	HTTP Monitoring Interface is updated to include Extended Services (XS) statistics	February 5, 2019
8.1.103.35	Treating incoming calls as inbound calls	February 5, 2019
8.1.103.25	HTTP Monitoring Interface is updated to include SIP Feature Server statistics	November 6, 2018
8.1.103.08	Secure SIP Signaling	May 15, 2018
	Migrating from Network SIP	December 13, 2017

Released in Version	Feature Name/Article	Date Released
	Server to SIP Server	
8.1.102.93	Remote agents with non-provisioned phone numbers	November 14, 2017
8.1.102.58	Enhanced procedure for upgrading SIP Server HA pair	February 21, 2017
8.1.102.51	Masking sensitive data in SIP messages	November 15, 2016
8.1.102.50	SRV address support in Contact and Record-Route headers	November 9, 2016
8.1.102.39	Metadata support for IVR recording	August 29, 2016
8.1.102.38	No-Answer Supervision: After Routing Timeout Action support in multi-site deployments	August 23, 2016
8.1.102.33	Overload Control: Logging Level	July 26, 2016
8.1.102.31	Customizing Music on Hold	July 8, 2016
8.1.102.29	Setting SIP INVITE timeout for individual DNs	June 28, 2016
8.1.102.28	HTTP Live Streaming	June 22, 2016
8.1.102.26	Recording an Agent Greeting	June 7, 2016
8.1.102.25	Controlling Early Media with a Routing Strategy	May 24, 2016
8.1.102.22	Dial Plan enhancements including support for SIP Feature Server Dial Plan	May 3, 2016
8.1.102.20	Muting/Unmuting a Party in Two-way Calls	April 26, 2016
8.1.102.13	HTTP Monitoring Interface	March 29, 2016
8.1.102.13	Improved presentation of multiple routing attempts in historical reporting	March 29, 2016
8.1.102.09	Enhanced Reporting for Early ISCC Transaction Completion	March 8, 2016
8.1.102.05	Modifying the From Header in SIP INVITE	February 8, 2016
8.1.102.02	Muting/Unmuting a Party in a Conference	January 19, 2016
8.1.102.01	No-Answer Supervision: After Routing Timeout Action	December 29, 2015
8.1.102.00	Caller Information Delivery Content for AT&T Trunks for GVP	December 18, 2015
8.1.101.97	Instant Messaging in Business Continuity deployments	December 8, 2015
8.1.101.95	DTMF Clamping in a Conference	November 24, 2015

Released in Version	Feature Name/Article	Date Released
	support in multi-site deployments	
8.1.101.87	Enable Customer-on-Hold Privacy	September 29, 2015
8.1.101.85	Providing Origination DN Name and Location in EventRinging	September 16, 2015
8.1.101.83	Sending Outgoing INVITEs with Multipart Body	September 9, 2015
8.1.101.78	Private Conversations During Conference	August 7, 2015
8.1.101.75	Alternate Routing for unresponsive URS/ORS	July 24, 2015
8.1.101.75	Find Me Follow Me	July 24, 2015
8.1.101.75	Shared Call Appearance support in Business Continuity deployments	July 24, 2015
8.1.101.74	Informing Agents of Supervisor Presence	July 8, 2015
8.1.101.68	DTMF Clamping in a Conference	June 8, 2015
8.1.101.66	Caller Information Delivery Content for AT&T Trunks for URS/ORS	May 21, 2015
8.1.101.62	Geo-location for MSML-based services: Strict Matching enhancement	March 30, 2015
8.1.101.62	Geo-location support by GVP enhancement	March 30, 2015
8.1.101.61	Nailed-up connection on agent login or when an agent in Ready state	March 17, 2015
8.1.101.57	Shared Call Appearance	February 12, 2015
8.1.101.57	Deleting Party From Conference	February 12, 2015
8.1.101.57	VXML support for agent greeting in multi-site and Business Continuity deployments	February 12, 2015
8.1.101.56	Agent Login and State Update on SIP Phones	January 30, 2015
8.1.101.50	Disabling Media Before Greeting	November 11, 2014
8.1.101.49	Hunt Groups support in Business Continuity deployments	October 31, 2014
8.1.101.47	Geo-location for MSML-based services: Strict Matching	October 21, 2014
8.1.101.43	Keep Alive for TCP Connections	September 19, 2014
8.1.101.38	Switching Between Supervision Modes	August 6, 2014
8.1.101.29	VXML support for agent greeting	June 20, 2014

Released in Version	Feature Name/Article	Date Released
8.1.101.29	Nailed-up connections in Business Continuity deployments	June 20, 2014
8.1.101.27	Hunt Groups enhancement to support sequential ringing	June 4, 2014
8.1.101.20	IMS Integration: Routed calls as originating or terminating	April 21, 2014
8.1.101.15	DN Recording Override	February 21, 2014
8.1.101.15	Dial Plan Support for Overdial	February 21, 2014
8.1.101.10	Trunk Capacity Control	January 31, 2014
8.1.101.10	Video Blocking	January 31, 2014