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SIP Server Deployment Guide

Improved presentation of multiple routing attempts in historical reporting

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With this enhancement, SIP Server provides additional information for historical reporting for the environment where **divert-on-ringing** is set to false and for call flows when a call is not answered by the first agent and must be distributed from a Routing Point multiple times.

In single-site routing scenarios, SIP Server generates EventReleased with AttributeCallState=7 (NoAnswer) for an unanswered party when the **after-routing-timeout** expires. Previously, SIP Server generated EventReleased with AttributeCallState=22 (Redirected).

To enable this feature in multi-site deployments, set the **sip-server-inter-trunk** configuration option to true on the Trunk DNs that are used to connect SIP Servers on different sites.

To retain the previous SIP Server behavior (prior to version 8.1.102.13), set the Application-level **enable-legacy-reporting** configuration option to true.

enable-legacy-reporting

Setting: TServer section, Application level

Default Value: false Valid Values: true, false

Changes Take Effect: On the next call

Enables backward compatibility for reporting AttributeCallState that SIP Server distributes in EventReleased for an unanswered routing target party in single-site routing scenarios.

- If set to true, SIP Server distributes EventReleased with AttributeCallState=22 (Redirected).
- If set to false, SIP Server distributes EventReleased with AttributeCallState=7 (NoAnswer).

Feature Limitations

In multi-site deployments, this feature works only with the **route** ISCC transaction type.