



# GE0-807<sup>Q&As</sup>

Genesys Certified Professional 8 - System Consultant, Outbound Voice

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### QUESTION 1

Which of the following options can be configured on a trunk DN to enable Active Out-of- Service detection? (Choose 2 answers)

- A. recovery-timeout\\'
- B. oos-check\\'
- C. \\'oos-force\\'
- D. \\'priority\\'

Correct Answer: BC

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### QUESTION 2

Which of the following events occur when SIP Server receives INFO and UPDATE messages?

- A. EventRouteRequest
- B. EventAddressInfo
- C. EventEstablished
- D. EventAttachedDataChanged

Correct Answer: A

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### QUESTION 3

What is the meaning of the option '\\sip-port\\' in the Tserver" section in '\\options\\' tab of the SIP server?

- A. It is the port used by SIP Server to communicate with other T-lib clients such as URS, StatServer, etc
- B. It is the port used by SIP Server to communicate with other SIP devices such as Media Server, SIP Endpoints, Gateways etc. using SIP signaling only
- C. It is the port used by SIP Server to communicate using SNMP protocol with a NMS software
- D. It is the port used by SIP Server to communicate with other SIP devices such as Media Server, SIP Endpoints, gateways etc. using RTP/RTCP protocol
- E. It is the port used by SIP Server to communicate with other SIP devices such as Media Server, SIP Endpoints,
- F. Gateways etc. using both RTP/RTCP protocol and SIP signaling.

Correct Answer: A

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**QUESTION 4**

You would like to record what the customer hears and says from the very beginning of the call to when it is terminated, regardless of whether or not an agent transferred the call, or created a conference, and so on. Which ON type should you use to enable recording?

- A. Extension
- B. RP
- C. Trunk
- D. VOIP port

Correct Answer: A

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**QUESTION 5**

Which statements are correct? (Choose 2 answers)

- A. It is possible to deploy a combination of standard TDM PBX(s) and sip server(s).
- B. Genesys SIP server is capable of processing SIP signing and RTP/RTCP streams as well.
- C. Customers who have a legacy PBX deployed and want to deploy Genesys SIP Server have to dismantle their legacy infrastructure.
- D. SIP Server supports the exchange of early media before a particular session is accepted.

Correct Answer: BD

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