

GE0-807^{Q&As}

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QUESTION 1

Which of the following is a correct description of how Media Server or Stream Manager can be configured with SIP Server in a multi-tenant environment?

- A. One Stream Manager can be associated with only one SIP Server
- B. One Stream Manager can be associated with multiple SIP Servers only under the same Tenant
- C. One Stream Manager can be associated with multiple SIP Servers and Tenants

Correct Answer: A

QUESTION 2

Which statements are correct? (Choose 2 answers)

- A. You need to install a Media Server In order to use Instant Messaging to play an Announcement.
- B. SIP Server supports the Integration with Genesys Media Server with both NETANN and MSML.
- C. Media Server can, if needed, convert a codec to another format.
- D. Genesys Media Server is based on the MGCP protocol for enhanced media functionality.

Correct Answer: BC

QUESTION 3

You have enabled the internal registrar. However, SIP Server does not update the key called contact of the corresponding DN when the endpoint successfully registers with SIP Server. Which two actions are required to address this issue? (Choose 2 answers)

- A. Set the option `external-registrar` to false
- B. Set the option `use-register-for-service-state` to true
- C. Set the `internal-registrar-persistent` to true
- D. Grant `change-permissions` for the System account for the all DNs on the corresponding switch

Correct Answer: CD

QUESTION 4

Which Genesys component is used to load balance between multiple SIP Servers?

- A. Load Distribution Server

- B. SIP Server
- C. Network SIP server
- D. Media Server

Correct Answer: C

QUESTION 5

A "Cradle to grave" ("beginning to end") call control is a functionality of which of the following? (Choose 2 answers).

- A. Registrar
- B. Stateless Proxy
- C. Stateful Proxy
- D. B2BUA
- E. Redirect Server

Correct Answer: AD

QUESTION 6

Sip Server 8.x introduced the overload control feature, which allows the specification of a value to limit the number of interactions SIP Server handles. At what percentage of this value does SIP Server reject all traffic?

- A. 101%
- B. 120%
- C. 130%
- D. 150%

Correct Answer: B

QUESTION 7

Which of the following are SIP messages that SIP Server can map to T-Library messages? Choose 4 answers

- A. ACK
- B. INVITE
- C. INFO
- D. UPDATE

E. REFER

F. ANNC

Correct Answer: BCDE

QUESTION 8

When SIP Server is behind the Softswitch, a DN is considered out of service if it receives which of the following messages?

A. 408 Request Timeout

B. 404 Busy

C. 482 Gone

D. 486 Forbidden

Correct Answer: A

QUESTION 9

SIP Server supports active out-of-service detection for which of the following types of ONs? (Choose 2 answers)

A. VoIP Service (MCU, treatment, etc.)

B. Extension

C. Trunk

D. Route Point

Correct Answer: AB

QUESTION 10

You want to install one SIP Server and 3 Media Servers. You want to communicate with them directly using NET ANN. You also want each Media Server to support MOH, treatment and MCU. What will be the minimum number of VoIP Service DNs you have to configure to support this request?

A. 1

B. 3

C. 6

D. 9

Correct Answer: A

QUESTION 11

Which of the following are SIP methods that can be used to pass DTMF tones? (Choose 3 answers)

- A. In-band method by encoding DTMF tone as regular RTP packets
- B. In-band method by encoding DTMF tone as in specific RTP packets (RFC 2833)
- C. Out-of-band method by encoding DTMF tone in SIP message (UPDATE method)
- D. Out-of-band method by encoding DTMF tone in SIP message (MESSAGE method)
- E. Out-of-band method by encoding DTMF tone in SIP message (INFO method)

Correct Answer: BDE

QUESTION 12

What is the response to REFER when the REFER method is used?

- A. 200 OK
- B. 202 Accepted
- C. 183 session in Progress
- D. C ACK
- E. C NOTIFY

Correct Answer: A

QUESTION 13

Which of the following are valid configurations for Open standard SIP architecture? Choose 2 answers

- A. 1 SIP Server, 1 Softswitch
- B. 5 SIP Server, 1 Network SIP Server
- C. 1 SIP Server, 1 T-Server
- D. 3 SIP Server, 5 T-Server, 1 Cisco call Manager

Correct Answer: AC

QUESTION 14

What does 1PCC mean?

- A. All call handling actions (answer, hold, initiate transfer, etc.) are performed on the SIP endpoint
- B. All call handling actions (answer, hold, initiate transfer, etc.) are performed on the same PC (no Hardware endpoint can be used)
- C. A T-lib client such as Agent Desktop can control a call which is actually between other parties such as sip endpoints
- D. Both, SIP endpoint and T-lib client such as Agent Desktop has to be integrated in one application

Correct Answer: A

QUESTION 15

Which of the following SIP Server features requires an MCU? (Choose 5 answers)

- A. 3 PCC conference - Single Step
- B. 3 PCC conference - Two Step
- C. Call Supervision
- D. Push video
- E. Emergency recording
- F. voicemail
- G. Email

Correct Answer: ABDFG

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