



72300X^{Q&As}

Avaya Aura Communication Applications Support Exam

Pass Avaya 72300X Exam with 100% Guarantee

Free Download Real Questions & Answers **PDF** and **VCE** file from:

<https://www.passapply.com/72300x.html>

100% Passing Guarantee
100% Money Back Assurance

Following Questions and Answers are all new published by Avaya
Official Exam Center

-  **Instant Download** After Purchase
-  **100% Money Back** Guarantee
-  **365 Days** Free Update
-  **800,000+** Satisfied Customers





QUESTION 1

In which two Avaya Aura® Communication Manager (CM) System Administration Terminal (SAT) forms does the SIP domain need to be configured? (Choose two.)

- A. signaling-group
- B. ip-network-region
- C. ip-network-map
- D. trunk-group

Correct Answer: AB

QUESTION 2

When a user is forwarded to Avaya Aura® Messaging (AAM), Avaya Aura® Communication Manager (CM) needs to indicate to AAM who the call was originally destined for, so that the message is left in the correct mailbox. Which SIP header is used to indicate for whom the call was originally destined, and the reason for being forwarded to AAM?

- A. B = history info
- B. To
- C. Request
- D. P-Asserted-Identity

Correct Answer: C

QUESTION 3

A maintenance engineer logs onto the Command Line Interface (CLI) of Avaya Aura® Messaging (AAM) and issues the vmstat command.

What information will the display present?

- A. The amount of virtual memory used
- B. The status information about the virtual machine
- C. The statistical information about voicemail boxes
- D. The count of mailbox password violation attempts

Correct Answer: A



QUESTION 4

Considering the message flow to an Avaya Aura® Contact Center (AACC) agent, which statement about the media path from the incoming caller is true?

- A. It is not anchored on the Avaya Aura® Media Server.
- B. It is anchored on the Avaya Aura® Media Server if using SIP agent.
- C. It is anchored on the Avaya Aura® Media Server until the agent answers the call.
- D. It is anchored on the Avaya Aura® Media Server for the duration of the call.

Correct Answer: B

QUESTION 5

Which tool has an option to debug the TLS handshake between the server and the client?

- A. SIP Trace Config and SIP Trace Viewer
- B. List trace tac 701
- C. traceSM
- D. List trace tac 701/tls

Correct Answer: B

[72300X Practice Test](#)

[72300X Study Guide](#)

[72300X Braindumps](#)