



300-815^{Q&As}

Implementing Cisco Advanced Call Control and Mobility Services
(CLACCM)

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QUESTION 1

Which two statements are correct with respect to the Client Matter Code setting in the route pattern configuration?
(Choose two.)

- A. The Client Matter Code feature does not support overlap sending because the Cisco Unified CM cannot determine when to prompt the user for the code.
- B. If you check the Allow Overlap Sending check box, the Require Client Matter Code check box becomes disabled.
- C. If you check the Allow Overlap Sending check box, you can also check the Require Client Matter Code check box.
- D. The Client Matter Code feature does support overlap sending because the Cisco Unified Communications Manager can determine when to prompt the user for the code.
- E. The Client Matter Code has the option to configure Authorization Level such as in the Forced Authorization Code.

Correct Answer: AB

QUESTION 2



The screenshot shows the Cisco Unified CM Administration interface for configuring a SIP Route Pattern. The page title is "Cisco Unified CM Administration For Cisco Unified Communications Solutions". The navigation menu includes System, Call Routing, Media Resources, Advanced Features, Device, Application, User Management, Bulk Administration, and Help. The main content area has a toolbar with Save, Delete, Copy, and Add New buttons. Below the toolbar is a "Status" section with an information icon. The "Pattern Definition" section contains the following fields: Pattern Usage* (Domain Routing), IPv4 Pattern* (*.cms.domain.local), IPv6 Pattern (empty), Description (empty), Route Partition (internal-pt), SIP Trunk/Route List* (CMS-Cluster-RL), and a Block Pattern checkbox. The "Calling Party Transformations" section includes Use Calling Party's External Phone Mask (checkbox), Calling Party Transformation Mask (empty), Prefix Digits (Outgoing Calls) (empty), Calling Line ID Presentation* (Default), and Calling Line Name Presentation* (Default). The "Connected Party Transformations" section includes Connected Line ID Presentation* (Default) and Connected Line Name Presentation* (Default). At the bottom, there are Save, Delete, Copy, and Add New buttons, and an information icon with the text "* - indicates required item."

Refer to the exhibit. The customer is troubleshooting an issue where users cannot dial the CMS SIP Route Pattern. The CMS URI they are attempting to dial is 7772002@cms.domain.local. Which IPv4 pattern must the customer enter to resolve the issue?

- A. 7772002@cms.domain.local
- B. *@cms.domain.local
- C. *.cms.domain.local
- D. cms.domain.local

Correct Answer: B

QUESTION 3



An engineer is troubleshooting Cisco Device Mobility and find that the phone has roamed to a building that is assigned to a different device pool but has not changed its device pool accordingly. What action resolves the issue?

- A. Set correct Location under Current Device Mobility Settings.
- B. Enable SRST under Current Device Mobility Settings.
- C. Set Device CSS under Current Device Mobility Settings.
- D. Set the correct subnet under Device Mobility info.

Correct Answer: D

QUESTION 4

A customer is using a SIP trunk to route calls to ITSP. To decrease the possibility of downtime, the customer invested in a failover device. How does the customer ensure reachability to ITSP, so that if one device on ITSP fails, the calls will be routed to another device?

- A. Enable SIP Option Ping on the SIP profile.
- B. Monitor the link using network management tools, and if it fails, manually change the routing to another working device.
- C. Enable ANAT on the SIP profile.
- D. Enable transmit security status on the SIP security profile.

Correct Answer: A

QUESTION 5

An administrator is working on an issue between the customer's Cisco Unified Border Element and the service provider. The provider only wants to see mid-call signaling from the Cisco Unified Border Element for fax calls. Which command must be configured on Cisco Unified Border Element?

- A. midcall-signaling passthru
- B. no update-callerid
- C. midcall-signaling passthru media-change
- D. midcall-signaling preserve-codec

Correct Answer: C