

300-815^{Q&As}

Implementing Cisco Advanced Call Control and Mobility Services
(CLACCM)

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QUESTION 1

A new deployment is using MVA for a specific user on the sales team, but the user is having issues when dialing DTMF. Which DTMF method must be configured in resolve the issue?

- A. in-band
- B. out-of-band
- C. gateway
- D. channel

Correct Answer: B

QUESTION 2

What are two configuration features of the Standard Local Route Group deployment? (Choose two.)

- A. is associated under the route group
- B. is associated only under the route list
- C. chooses the route group that is configured under the device pool of the calling-party device
- D. chooses the route group that is configured under the device pool of the called-party device
- E. is assigned directly to the route pattern

Correct Answer: BC

QUESTION 3

Cisco Unified Communications Manager Dialed Number Analyzer

DNA Analysis Output Save the Displayed Output

Cisco Unified Communications Manager Dialed Number Analyzer Results

▼ Results Summary

▼ Calling Party Information

- Calling Party = 4125551212
- Partition =
- Device CSS =
- Line CSS = ONNET-CSS
- AAR Group Name =
- AAR CSS =

● Dialed Digits = 914125550000

● Match Result = RouteThisPattern

▼ Matched Pattern Information

- Pattern = 9.1[2-9]XX[2-9]XXXXXX
- Partition =
- Time Schedule =

● Called Party Number = 14125550000

● Time Zone = Etc/GMT

● End Device = HQ-RL

● Call Classification = OffNet

● InterDigit Timeout = NO

● Device Override = Disabled

● Outside Dial Tone = NO

▶ Call Flow

▶ Alternate Matches

NOTE: The analysis results are purely based on configurations available in the Cisco Communications Manager database. For Gateway outbound calls, call details might differ depending on the Gateway's settings.

▼ Call Flow

▼ Route Pattern :Pattern= 9.1[2-9]XX[2-9]XXXXXX

- Positional Match List =
- DialPlan =

▼ Route Filter

- Filter Name =
- Filter Clause =

○ Require Forced Authorization Code= NO

○ Authorization Level= 0

○ Require Client Matter Code= NO

○ Call Classification =

○ PreTransform Calling Party Number = 4125551212

○ PreTransform Called Party Number = 914125550000

▼ Calling Party Transformations

- External Phone Number Mask = NO
- Calling Party Mask =
- Prefix =
- CallingLineId Presentation = Default
- CallingName Presentation = Default
- Calling Party Number = 4125551212

▼ ConnectedParty Transformations

- ConnectedLineId Presentation = Default
- ConnectedName Presentation = Default

▼ Called Party Transformations

- Called Party Mask =
- Discard Digits Instruction = PreDot
- Prefix =
- Called Number = 14125550000

▼ Route List :Route List Name= HQ-RL

- ▼ RouteGroup: RouteGroup Name= Standard Local Route Group
 - PreTransform Calling Party Number = 4125551212
 - PreTransform Called Party Number = 914125550000
 - ▼ Calling Party Transformations
 - External Phone Number Mask = Default
 - Calling Party Mask =
 - Prefix =
 - Calling Party Number = 4125551212
 - ▼ Called Party Transformations
 - Called Party Mask =
 - Discard Digits Instruction =
 - Prefix =
 - Called Number = 914125550000

▶ Alternate Matches

- Note: Information Not Available

Refer to the exhibit. A collaboration engineer is troubleshooting an issue where the PSTN calls of a Cisco UCM IP phone user are not reaching the PSTN gateway. Which action resolves the issue?

- A. Change the calling search space of the user's line or device.
- B. Change the "Call Classification" to "OnNet" on the route pattern.
- C. Ensure that the user's phone is assigned to a device pool with the correct local route settings.
- D. Deselect "Block this pattern" on the route pattern.

Correct Answer: D

QUESTION 4

Which set of commands binds SIP media and signaling to interface GigabitEthernet0/0 when dial peer 1 is chosen for call routing?

- A. dial-peer voice 1 voip voice-class source interface GigabitEthernet0/0
- B. voice service voip bind sip source-interface GigabitEthernet0/0
- C. voice service voip sip bind all source-interface GigabitEthernet0/0
- D. dial-peer voice 1 voip voice-class bind control source-interface GigabitEthernet0/0 voice-class sip bind media source-interface GigabitEthernet0/0

Correct Answer: C

QUESTION 5

The sales department must answer phones when other sales members are not at their desks. The administrator knows that configuring Call Pickup allows the sales users to answer all the calls in the department by pressing only the softkey. Which call pickup configuration meets this requirement?

- A. Standard Call Pickup
- B. Group Call Pickup
- C. Other Group Call Pickup
- D. Directed Call Pickup

Correct Answer: B

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