



# 300-815<sup>Q&As</sup>

Implementing Cisco Advanced Call Control and Mobility Services  
(CLACCM)

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## QUESTION 1

A customer routes PSTN calls to ITSP through a SIP trunk on Cisco UCM that forwards and receives calls to and from ITSP. ITSP is set to send an E.164 number when the customer's extension is four digits. Which action should be taken to route the incoming calls to four-digit extensions?

- A. Configure a voice translation profile to map the E.164 number to four digits and assign it to the incoming dial-peer on Cisco Unified Border Element.
- B. Set the Significant Digits to 8 on the SIP trunk.
- C. Set the Significant Digits to 4 on the SIP trunk.
- D. Configure a voice translation rule to map the E.164 number to four digits and assign it to the incoming dial-peer on Cisco Unified Border Element.

Correct Answer: C

## QUESTION 2

The screenshot displays the Cisco Unified CM Administration interface. On the left, the 'Find and List Route Patterns' page shows a table of route patterns. The table has columns for Pattern, Description, Partition, Route Filter, Associated Device, and Copy. The first row shows a pattern '9.011' with description 'PSTN-PT' and partition 'SIP-PT'. The second row shows '9.011-1' with description 'PSTN-PT' and partition 'SIP-PT'. The third row shows '9.011-1000000' with description 'PSTN-PT' and partition 'SIP-PT'. The fourth row shows '9.011-1000000' with description 'PSTN-PT' and partition 'SIP-PT'. The fifth row shows '141' with description 'PSTN-PT' and partition 'SIP-PT'. On the right, the 'Cisco Unified Communications Manager Dialed Number Analyzer' page shows the 'DINA Analysis Output' for the pattern '9.011-1000000'. The output includes a 'Results Summary' section with 'Calling Party Information' (Calling Party: 4125551212, Partition: SIP-PT, Line CSS: DTMF-CSS, AAR Group Name: SIP-PT, AAR CSS: SIP-PT), 'Matched Digits' (9.011-1000000), 'Match Result' (BlockThisPattern), 'Route Block Cause' (No route found), 'Called Party Number' (4125551212), 'Matched Pattern Information' (Pattern: 9.011-1000000, Partition: SIP-PT, Pattern Type: SIP/CHT, Time Zone: PST/PST, Outside Dial Tone: NO), 'Call Flow' (Note: Information Not Available), and 'Alternate Matches' (Note: Information Not Available).

Refer to the exhibit. A collaboration engineer is troubleshooting an issue where a user of a Cisco UCM IP phone reports failed calls when trying to dial out to the PSTN. Which action resolves the issue?

- A. Assign a calling search space to the line or the device that has access to the route pattern.
- B. Deselect "Block this pattern" on the "Route Option" setting of the route pattern.
- C. Select the "Urgent Priority" setting on the route pattern.
- D. Instruct the user to not dial a "1" before their local area code.

Correct Answer: A



### QUESTION 3

A company was looking at the IT charges and saw many long-distance and international calls primarily to sites in North America and around the world. The administrator wants to optimize the PSTN expense. Which dial plan configuration reduces PSTN connectivity charges by using the IP network to bring the egress point to the PSTN as close as possible to the called number?

- A. translation patterns
- B. tail end hop off
- C. client matter codes
- D. dial rules

Correct Answer: B

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### QUESTION 4

An administrator is configuring a new deployment using Cisco Unified CME. The SCCP phones register without any issues, but SIP phones are not registering. Assume that all other configuration is valid. Which code allows SIP phones to register to Cisco UCME?

- A. voice service voip allow-connections sip to h323
- B. voice service voip sip bind media source-interface Vlan100
- C. voice service voip sip bind control source-interface Vlan100
- D. voice service voip sip registrar server expires max 600 min 60

Correct Answer: D

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### QUESTION 5

A company is using Cisco Jabber on-premises to make B2B calls on video. The calls are using Cisco Expressway-C and Expressway-E and have been configured in Cisco UCM to be able to call any URI on the internet. The Jabber client also

has voice enabled and must be able to call local, regional, and international numbers.

Where must Cisco UCM be configured to meet this requirement for URI dialing?

- A. Enter “!#” in the SIP route pattern.



B. Enter “.” in the route pattern section tied to a route group and list.

C. Enter “\*” in the SIP route pattern.

D. Enter “!#” in the route pattern section tied to a route group and list.

Correct Answer: B

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