



300-815^{Q&As}

Implementing Cisco Advanced Call Control and Mobility Services
(CLACCM)

Pass Cisco 300-815 Exam with 100% Guarantee

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

<https://www.passapply.com/300-815.html>

100% Passing Guarantee
100% Money Back Assurance

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

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



**QUESTION 1**

A company has users that are logged in to hunt groups. However, there is a requirement for hunt group configurations to provide an option to turn on audible ringtones when calls to a line group arrive at a phone that is logged out and on a break. This ringtone alerts a logged-out user that there is an incoming call to a hunt list to which the line is a member, but the call does not ring at the phone of that line group member because of the logged-out status. Which action meets this requirement?

- A. Configure the HLog softkey on the phone so that while a user is logged off, it plays an audible tone when a call is missed.
- B. Set the service parameter Enterprise Feature Access number for hunt group logout and set up an access number.
- C. Set the service parameter Party Entrance Tone to "True."
- D. Configure the service parameter hunt group logoff notification and specify the name of the ringtone file.

Correct Answer: D

QUESTION 2

A company is experiencing an issue where calls from the Cisco UCM via Cisco Unified Border Element to the ITSP fail with a "500 server internal error" right after a 183 session in progress is sent back from the ITSP. What must be enabled to resolve the issue?

- A. delayed offer on Cisco UCM
- B. early offer at the CUBE
- C. delayed offer on CUBE
- D. early offer at the ITSP

Correct Answer: B

QUESTION 3

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

QUESTION 4

Refer to the exhibit.

```
SIP/2.0 183 Session Progress
Via: SIP/2.0/UDP 192.168.100.100:5060
From: <sip:+123456789@192.168.100.100>;
To: <sip:987654321@192.168.100.200>
Date: Fri, 28 Jun 2019 08:30:32 GMT
Call-ID: fce8c980-d151d028-19cf3-325900a@192.168.100.100
CSeq: 101 INVITE
Require: 100rel
RSeq: 101
Allow: INVITE, OPTIONS, BYE, CANCEL, ACK, PRACK, UPDATE, REFER, SUBSCRIBE, NOTIFY, INFO, REGISTER
Contact: <sip:987654321@192.168.100.200:5060>
Content-Type: application/sdp
Content-Disposition: session;handling=required
Content-Length: 247

v=0
o=CiscoSystemsSIP-GW-UserAgent 4780 5245 IN IP4 192.168.100.200
s=SIP Call
c=IN IP4 192.168.100.200
t=0 0
m=audio 16384 RTP/AVP 8 101
c=IN IP4 192.168.100.200
a=rtpmap:8 PCMA/8000
a=rtpmap:101 telephone-event/8000
a=fmtp:101 0-15
a=ptime:20
```

While troubleshooting call failures on the Cisco Unified Border Element, an administrator notices that messages are being sent to the service provide, but there is no response. The administrator later learns that this SIP provider does not support PRACK. Which header should be removed from the SIP message to resolve this issue?

- A. Require: 100rel
- B. Content-Type: application/sdp
- C. Contact:
- D. Content-Disposition: session;handling=required

Correct Answer: A

QUESTION 5



Cisco Unified Communications Manager Dialed Number Analyzer

DNA Analysis Output

Save the Displayed Output

Save

Cisco Unified Communications Manager Dialed Number Analyzer Results

Expand All

Collapse All

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

300-815 VCE Dumps | 300-815 Practice Test | 300-815 Exam Questions

4 / 5



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

[300-815 VCE Dumps](#)

[300-815 Practice Test](#)

[300-815 Exam Questions](#)