

# 300-815 Free Sample Questions, Authorized 300-815 Test Dumps



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Cisco 300-815 exam is composed of 60-70 questions and has a time limit of 90 minutes. 300-815 exam covers a wide range of topics such as configuring and troubleshooting Cisco Unified Communications Manager (CUCM), implementing call routing and dial plans, integrating Cisco Unity Connection and Cisco Unity Express, and deploying mobility features such as Mobile and Remote Access (MRA) and Cisco Jabber. Candidates are expected to have a solid understanding of these topics, as well as practical experience with Cisco collaboration solutions.

The Cisco 300-815 Exam is intended for IT professionals who are responsible for designing, implementing, and managing advanced call control and mobility services in a Cisco environment. 300-815 exam covers a wide range of topics, including Cisco Unified Communications Manager (CUCM), Cisco Unified Communications Manager Express (CUCME), Cisco Unity Connection, Cisco Unified IM and Presence, and Cisco Expressway.

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## Free PDF Cisco - 300-815 Updated Free Sample Questions

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## Cisco Implementing Cisco Advanced Call Control and Mobility Services

## Sample Questions (Q149-Q154):

### NEW QUESTION # 149

An engineer is configuring a Cisco UCM solution. The requirements state that some users in one location will receive calls from a number during work hours 09AM to 5PM, and another group will get the calls from the same number outside this defined timeslot. Users will also change the outgoing number when reaching out to customers based on the same time-of-day routing rules. Which feature is needed to allow for this type of configuration?

- A. CSS
- B. partitions
- C. route groups
- D. route lists

**Answer: B**

### NEW QUESTION # 150

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?

**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**

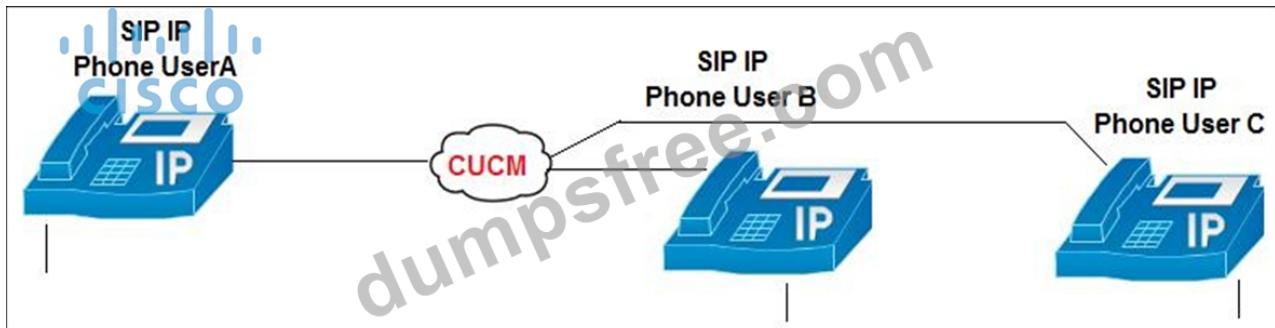
- Note: Information Not Available

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

Answer: D

#### NEW QUESTION # 151

Refer to the exhibit.



In an active SIP call between phone user A and phone user B, phone A initiates a call transfer to phone user C. Which two scenarios are correct? (Choose two.)

- A. As soon as Phone\_A presses the Transfer button for the first time, Phone\_B hears the MOH and the MOH audio is chosen from Phone\_A User Hold MOH Audio Source settings.
- B. As soon as Phone\_A presses the Transfer button for the first time, Phone\_B hears the MOH and the MOH audio is chosen from Phone\_B User Hold MOH Audio Source settings.
- C. As soon as Phone\_A presses the Transfer button for the first time, Phone\_B hears the music on hold and the MOH audio is chosen from Phone\_A Network Hold MOH Audio Source settings.
- D. Phone\_A sends a SIP-REFER message to the Cisco Unified Communications Manager with Phone\_C information in the Refer-To section.
- E. Phone\_B sends a SIP-REFER message to the Cisco Unified CM with Phone\_C information in the Refer-To section.

**Answer: C,D**

**NEW QUESTION # 152**

An administrator is troubleshooting call failures on an H.323 gateway via the CLI. To see signaling for media and call setup, which debug must the Administrator turn on? (Choose two)

- A. debug H.323 asn1
- B. **debug H.225 asn1**
- C. **debug H.245 asn1**
- D. debug H.225 media
- E. debug H.323 messages

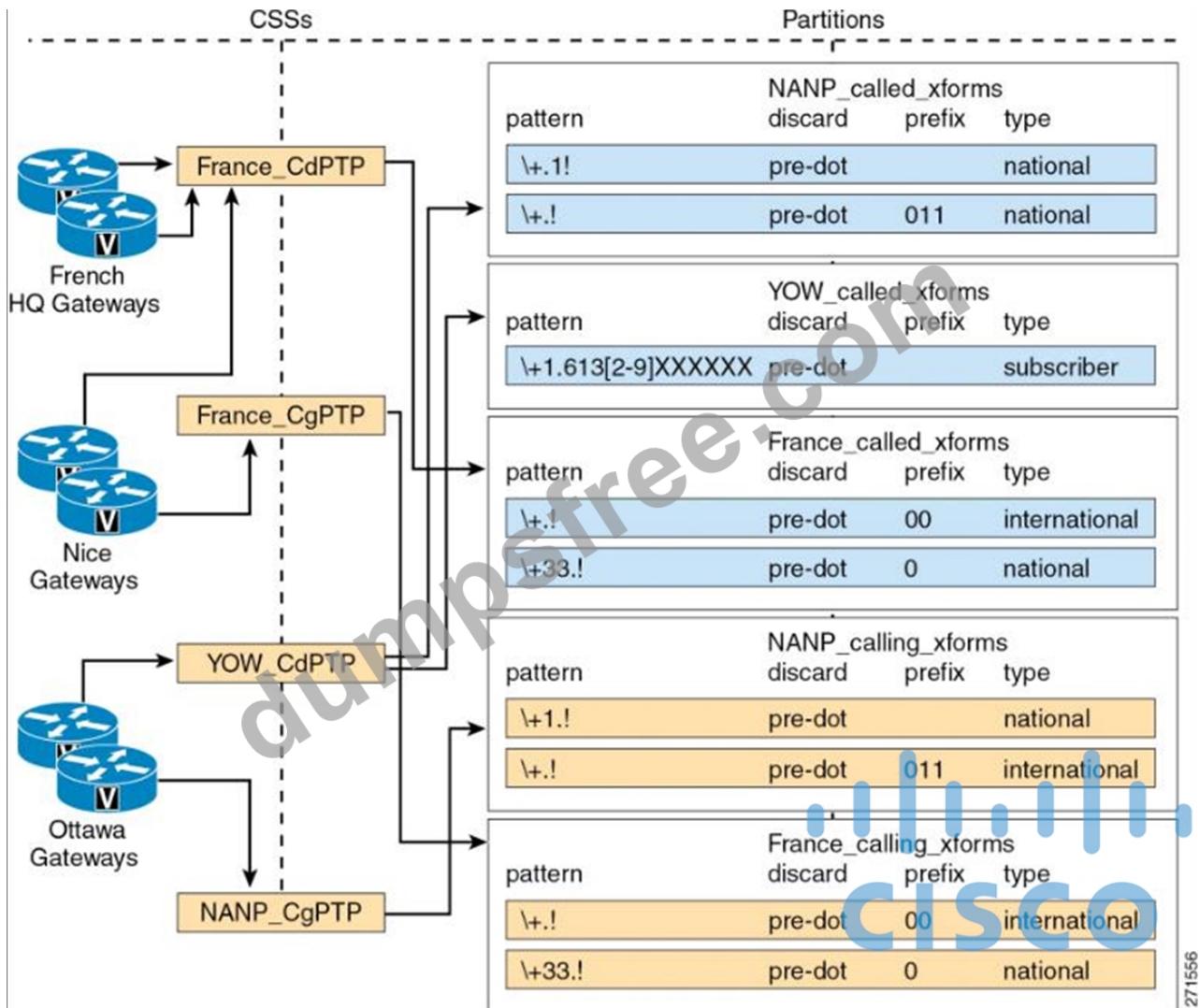
**Answer: B,C**

Explanation:

<https://www.ccexpert.us/ip-telephony/selected-debug-commands.html>

**NEW QUESTION # 153**

Refer to the exhibit.



Within the North American Numbering Plan, gateways located in Ottawa, Canada and marked as "YOW" are assigned to the Calling Party Transformation CSS NANP\_CgPTP, which contains partition NANP\_calling\_xforms. What is the calling-party number and the numbering type if the calling user +1613-555-1234 dials the number?

- A. calling number 011-1-613-555-1234 and numbering type "subscriber"
- B. calling number 011613-555-1234 and numbering type "international"
- C. calling number 613-555-1234 and numbering type "subscriber"
- D. calling number 613-555-1234 and numbering type "national"**

**Answer: D**

#### NEW QUESTION # 154

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