350-801^{Q&As}

Implementing and Operating Cisco Collaboration Core Technologies (CLCOR)

Pass Cisco 350-801 Exam with 100% Guarantee

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

https://www.leads4pass.com/350-801.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

Leads4Pass

800,000+ Satisfied Customers



Leads4Pass

QUESTION 1

Refer to the exhibit.

REGION A	REGION A-B Audio Codec Preference List:	REGION B	
	(1).G722	· - •	Sip Early Offer Trun
	(2).G711 (3).G729	¥∎€	SIP INVITE Codec Preference In Early Offer SDP
Codecs supported: G722 G711			
G729	Region A-B Max Audio Bitrate: 40kbps		

Unanswered calls do not reach the voicemail associated with the phones. Instead, callers receive the default greeting. Which action fixes the configuration?

- A. Review the conversation manager logs on Cisco Unity Connection.
- B. Check the box "Redirecting Diversion Header Delivery Outbound", then reset the trunk.
- C. Reboot Cisco Unity Connection.
- D. Check the box "Redirecting Diversion Header Delivery Outbound".

Correct Answer: B

QUESTION 2

A customer wants to deploy Cisco Jabber Phone Mode with contacts. Which two configurations are needed to accomplish this goal? (Choose two)

- A. Add the Application Client Users role to the Cisco UCM end users.
- B. Remove the IM and Presence servers from the Cisco UCM service profile.
- C. Disable Instant Messaging on IM and Presence.
- D. Import Contacts.xml into Jabber.
- E. Enable Cisco UCM end-users for IM and Presence.

Correct Answer: CE

Leads4Pass

https://www.cisco.com/c/en/us/support/docs/unified-communications/jabber/213403-configure-jabber-version-12-0-in-phone-m.html

QUESTION 3

What are two characteristics of priority queueing? (Choose two.)

- A. It guarantees bandwidth for user-defined traffic classes.
- B. It applies strict priority to user-defined traffic classes.
- C. It allocates an equal share of bandwidth to each flow.
- D. It places traffic Into one of four queues.
- E. It ensures that critical traffic queues are serviced first.

Correct Answer: DE

QUESTION 4

How does an administrator make a Cisco IP phone display the last 10 digits of the calling number when the call is in the connected state, and also display the calling number in the E.164 format within call history on the phone?

A. Configure a translation pattern that has a Calling Party Transform Mask of XXXXXXXXXX.

B. On the inbound SIP trunk, change Significant Digits to 10.

- C. Change the service parameter Apply Transformations On Remote Number to True.
- D. Configure a calling party transformation pattern that keeps only the last 10 digits.

Correct Answer: D

Translationpattern might work, but have to be triggered as called number equals directory number, not suitable sip-trung can change digits on calling number Calling Party Transformations on IP Phones Calling Party Transformation Patterns allow the system to adapt the calling party numbers to different formats. The most typical use is to adapt from globalized to localized calling party numbers and vice versa.

https://www.cisco.com/c/en/us/td/docs/voice_ip_comm/cucm/srnd/collab11/collab11/dialplan.html

QUESTION 5

Which configuration step is necessary for a Cisco SIP phone to synchronize its time with a specific source?

A. Add a Phone NTP Reference to the Date/Time Group.

- B. Assign the device to the correct region.
- C. Change the Time Format from 24-hour to 12-hour.
- D. Change the Time Zone from "America/Los_Angeles" to "Etc/GMT+8".

Leads4Pass

Correct Answer: A

Anything other than A is a situational preference, or just plain wrong.

If you\\'re in the east coast of the US, why would it be a requirement to set your time zone to LA?

It doesn\\'t matter whether you\\'re using 12 or 24 hour format.

Assigning a device to a region doesn\\'t change the time source for that device - if anything that\\'s handled by Device Mobility.

-

The only sensible solution is to add the Phone NTP source to the Date/Time group.

Reference: https://www.cisco.com/c/en/us/td/docs/voice_ip_comm/cucm/admin/10_0_1/ccmcfg/CUCM_BK_C95ABA82_00_admin-guide-100/CUCM_BK_C95ABA82_00_admin-guide-100_chapter_0110.html

Latest 350-801 Dumps

350-801 Practice Test

350-801 Study Guide