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## Cisco Implementing and Operating Cisco Collaboration Core Technologies Sample Questions (Q123-Q128):

### NEW QUESTION # 123

An administrator needs to stop the leading 9 from outbound calls on an IOS Voice Gateway and ensure that the system handles 911 emergency calls.

Which configuration is needed to accomplish this task?

- A. 

```
voice translation-rule 1
rule 1 /9?911/ /911/
rule 2 /^9\(.*\)/ /&/
```
- B. 

```
voice translation-rule 1
rule 1 /^9\(.*\)/ /\1/
rule 2 /9?911/ /911/
```
- C. 

```
voice translation-rule 1
rule 1 /9?911/ /911/
rule 2 /^9\(.*\)/ /\0/
```
- D. 

```
voice translation-rule 1
rule 1 /9?911/ /911/
rule 2 /^9\(.*\)/ /1/
```

Answer: D

### NEW QUESTION # 124

A collaboration engineer must configure Cisco Unified Border Element to support up to five concurrent outbound calls across an Ethernet Link with a bandwidth of 160 kb to the Internet Telephony service provider.

Which set of commands allows the engineer to complete the task without compromising voice quality?

- A) 

```
dial-peer voice 1 voip
translation-profile outgoing Strip9
max-conn 5
destination-pattern 91[2-9]-[2-9]...$
session protocol sip2
session target ipv4:142.45.10.1
dtmf-relay rtp-nte sip-notify sip-kpml
codec aacld
```
- B) 

```
dial-peer voice 1 voip
translation-profile outgoing Strip9
max-conn 5
destination-pattern 91[2-9]-[2-9]...$
session protocol sip2
session target ipv4:142.45.10.1
dtmf-relay rtp-nte sip-notify sip-kpml
codec mp4a-ltm
```
- C) 

```
dial-peer voice 1 voip
translation-profile outgoing Strip9
max-conn 5
destination-pattern 91[2-9]-[2-9]...$
session protocol sip2
session target ipv4:142.45.10.1
dtmf-relay rtp-nte sip-notify sip-kpml
codec mp4a-ltm
```
- D) 

```
dial-peer voice 1 voip
translation-profile outgoing Strip9
max-conn 5
destination-pattern 91[2-9]-[2-9]...$
session protocol sip2
session target ipv4:142.45.10.1
dtmf-relay rtp-nte sip-notify sip-kpml
```

- A. Option B
- B. Option C
- C. Option A
- D. Option D

**Answer: A**

#### NEW QUESTION # 125

Which task is required when configuring self-provisioning for an end user in Cisco UCM?

- A. Disable Auto-Registration.
- B. Associate the end user to a SIP Profile.
- C. Enable Auto-Registration.
- D. Associate the end user to the Standard CCM Super Users group.

**Answer: C**

#### NEW QUESTION # 126

An engineer must add resources to a Cisco Catalyst gateway that is currently used to route calls to and from the PSTN to provide these additional services:

- conferencing
- transcoding

What must be added to the gateway?

- A. hardware transcoder
- B. MTP
- C. digital signal processor
- D. Cisco IP Voice Media Streaming application

**Answer: C**

Explanation:

Digital Signal Processors (DSPs) are specialized hardware components that handle media processing tasks such as voice compression, echo cancellation, conferencing, and transcoding.

By adding DSPs to your Cisco Catalyst gateway, you provide the necessary resources to support these services.

Reference:

[https://www.cisco.com/en/US/docs/voice\\_ip\\_comm/cucm/admin/3\\_3\\_3/ccmsys/a05dsp.html](https://www.cisco.com/en/US/docs/voice_ip_comm/cucm/admin/3_3_3/ccmsys/a05dsp.html)

#### NEW QUESTION # 127

Which Webex Calling construct is used to organize calling features within a physical site?

- A. locations
- B. service settings
- C. client settings
- D. call routing

**Answer: A**

Explanation:

A location is a physical site that contains users, devices, and resources. Locations are used to organize calling features within a physical site. For example, you can create a location for each of your offices and then assign users, devices, and resources to that location. This will allow you to manage calling features for each office separately.

#### NEW QUESTION # 128

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