

## Cisco.300-070.v2019-01-30.q181

|                                                                                                                                                         |                                                          |
|---------------------------------------------------------------------------------------------------------------------------------------------------------|----------------------------------------------------------|
| Exam Code:                                                                                                                                              | 300-070                                                  |
| Exam Name:                                                                                                                                              | Implementing Cisco IP Telephony & Video, Part 1 (CIPTV1) |
| Certification Provider:                                                                                                                                 | Cisco                                                    |
| Free Question Number:                                                                                                                                   | 181                                                      |
| Version:                                                                                                                                                | v2019-01-30                                              |
| # of views:                                                                                                                                             | 1079                                                     |
| # of Questions views:                                                                                                                                   | 49533                                                    |
| <a href="https://www.freecram.net/torrent/Cisco.300-070.v2019-01-30.q181.html">https://www.freecram.net/torrent/Cisco.300-070.v2019-01-30.q181.html</a> |                                                          |

### NEW QUESTION: 1

An engineer has hundreds of H.323 gateways with thousands of route patterns in a Cisco Unified Communications Manager cluster. The engineer has noticed many specific route patterns created to the H.323 gateways. Which digit manipulation method on route patterns is most effective in reducing the number of route patterns?

- A. regular expressions
- B. called-party transform mask
- C. prefix digits
- D. discard digits

Answer: ([SHOW ANSWER](#))

### NEW QUESTION: 2

How many TelePresence MCU conference bridges can be configured in one conference bridge pool in a TelePresence Conductor cluster?

- A. 24
- B. 30
- C. 1
- D. 12
- E. 36

Answer: ([SHOW ANSWER](#))

### NEW QUESTION: 3

If the route patterns 5XXX, 51XX, 513X and 5134 exist within a call route database, which pattern would be matched if the dialed number was 5324?

- A. 5134
- B. 513X
- C. 5XXX

D. 51XX

Answer: ([SHOW ANSWER](#))

**NEW QUESTION: 4**

An organization has built their video infrastructure to perform ad hoc video calls, but they also need to conduct managed video conference calls. Which hardware component is needed to conduct these calls?

- A. MCU
- B. VCS-E
- C. TelePresence server
- D. IOS gateway

Answer: ([SHOW ANSWER](#))

**NEW QUESTION: 5**

An organization with a centralized dial plan has user extensions as the last four digits of the external DID number. Some users at the head office have overlapping extensions with the users at the branch offices. Which action fixes this issue without changing user extensions and allowing users to call each other across sites?

- A. Use site codes.
- B. Use different CSSs for both sites.
- C. Place both extensions in different partitions.
- D. Change the number of digits in the internal extensions.

Answer: ([SHOW ANSWER](#))

**NEW QUESTION: 6**

Refer to the exhibit.

**Cisco Unified CM Administration**  
For Cisco Unified Communications Solutions

System ▾ Call Routing ▾ Media Resources ▾ Advanced Features ▾ Device ▾ Application ▾ User Management ▾ Bul

### Route List Configuration

Save 
 Delete 
 Copy 
 Reset 
 Apply Config 
 Add New

Registration: Registered with Cisco Unified Communications Manager 10.100.4.68  
 IP Address: 10.100.4.68  
☒ Device is trusted  
 Name\*: RL\_SLRG  
 Description: Route List Standard Local Route Group  
 Cisco Unified Communications Manager Group\*: CMG\_Daxa ▾  
☒ Enable this Route List (change effective on Save; no reset required)  
☐ Run On All Active Unified CM Nodes

---

#### Route List Member Information

Selected Groups\*\*: Standard Local Route Group  
 ▼ ▲ Add Route Group  
 ▼ ▲  
 Removed Groups\*\*\*:

---

#### Route List Details

[Standard Local Route Group](#)

What is the benefit of using the Standard Local Route Group?

- A. You can make call routing decisions based on the closest gateway to the destination of the call (that is, the TEHO).
- B. You can have a simplified dialing plan, in which you use the same route patterns, calling search spaces, and partitions for all offices in the same dialing domain (country), and you make call routing decisions based on the device pool of the calling device.
- C. You can have a noncentralized dialing plan for all company offices, to ensure reliability and remote survivability.
- D. The Standard Local Route Group is used exclusively for local calls.

Answer: ([SHOW ANSWER](#))

#### NEW QUESTION: 7

Which three feature types can be configured on Cisco Unified Border Element? (Choose three.)

- A. Cisco Unified Border Element does not support media flow-around
- B. contact center features such as call queuing, agent, and supervisor features
- C. automated attendant and voicemail features such as message waiting indication and notification
- D. session management features such as CAC and dial-plan interpretation and routing

- E. demarcation features such as topology hiding, statistics, and billing
- F. interworking features such as H.323-SIP interworking, SIP normalization, DTMF type conversion, and payload type conversion

**Answer:** ([SHOW ANSWER](#))

#### **NEW QUESTION: 8**

Which protocol is recommended to be used between Cisco Unified Communications Manager and the voice gateway to simplify the dial plan?

- A. RSVP
- B. H323
- C. MGCP
- D. SIP
- E. SCCP

**Answer:** C ([LEAVE A REPLY](#))

#### **NEW QUESTION: 9**

What is the relationship between a Region and a Location?

- A. The Region codec parameter is used between a Region and its configured Locations.
- B. The Region codec parameter is combined with Location bandwidth when communicating with other Regions.
- C. The codec parameter configured in the Region is only used between Regions and Location bandwidth is only used between Locations.
- D. The Region setting for a Location sets the number of audio and video calls that Location can support.

**Answer:** ([SHOW ANSWER](#))

#### **NEW QUESTION: 10**

Which Cisco Unified Communications Manager default global conference setting should be changed to ensure fraud prevention?

- A. Advanced Ad Hoc Conference Enabled
- B. Enable Click-to-Conference for Third-Party Applications
- C. Drop Ad Hoc Conference
- D. Block OffNet to OnNet Transfer
- E. MCU Conference Bridge Custom Layout Index

**Answer:** ([SHOW ANSWER](#))

Explanation/Reference:

Explanation:

Cisco Unified Communications Manager Administration provides the clusterwide service parameter, Drop Ad Hoc Conference, to allow the prevention of toll fraud (where an internal conference controller disconnects from the conference while outside callers remain connected). The service parameter settings specify conditions under which an ad hoc conference gets dropped.

Reference: [http://www.cisco.com/c/en/us/td/docs/voice\\_ip\\_comm/cucm/admin/8\\_6\\_1/ccmsys/accm-861-cm/a05confb.html](http://www.cisco.com/c/en/us/td/docs/voice_ip_comm/cucm/admin/8_6_1/ccmsys/accm-861-cm/a05confb.html)

#### NEW QUESTION: 11

Which two of these are required in order for Cisco Unified Communications Manager to support software conferencing? (Choose two.)

- A. The Cisco IP Voice Media Streaming Application needs to be activated on the server running the conferencing service in the cluster.
- B. Under the Cisco IP Voice Media Streaming Application parameters, the conference bridge needs to have Run Flag set to True.
- C. The service parameter for the Cisco IP Voice Media Streaming Application needs to have Run Flag set to True.
- D. The software conference bridge resource needs to be configured in Cisco Unified Communications Manager.
- E. The Cisco IP Voice Media Streaming Application needs to be activated for all servers in the cluster.

Answer: ([SHOW ANSWER](#))

#### NEW QUESTION: 12

How does Cisco Unified Communications Manager perform digit analysis?

- A. Perform a digit-by-digit analysis to find the best matching route pattern based on a devices partition.
- B. Send all collected digits through the Dialed Number Analyzer to determine the best call routing match within Cisco Unified CM.
- C. Perform a digit-by-digit analysis to find the best matching route pattern within the partition list in the calling search space of a device.
- D. Use a set of route lists and filters to match all collected digits of a dialed number.

Answer: ([SHOW ANSWER](#))

#### NEW QUESTION: 13

What is a route list?

- A. a list of locations
- B. an ordered list of regions
- C. a list of route patterns
- D. an ordered list of gateways and trunks
- E. an ordered list of route groups

Answer: ([SHOW ANSWER](#))

Explanation/Reference:

Reference: [http://www.cisco.com/c/en/us/td/docs/voice\\_ip\\_comm/cucm/admin/7\\_1\\_2/ccmsys/accm-712-cm/a03rp.html#wp1045311](http://www.cisco.com/c/en/us/td/docs/voice_ip_comm/cucm/admin/7_1_2/ccmsys/accm-712-cm/a03rp.html#wp1045311)

#### NEW QUESTION: 14

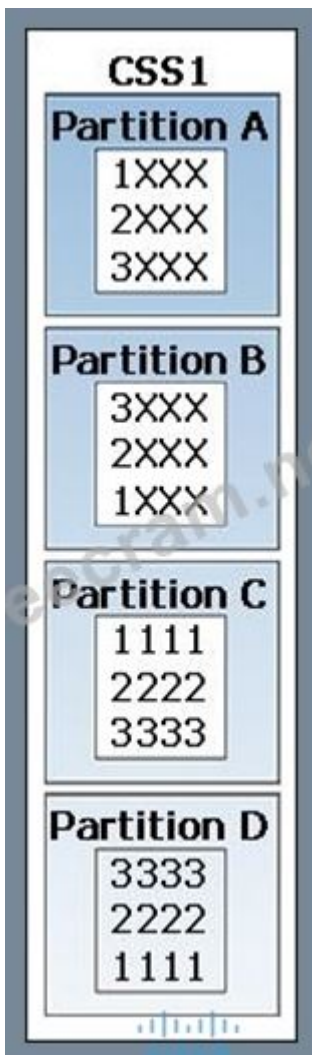
When configuring Cisco Unified Communications Manager to use Busy Lamp Field Speed Dials as a native presence feature, which function is required?

- A. Enable the "BLF for Call Lists" in the Cisco CallManager service parameter section.
- B. Enable the "BLF for Call Lists" in the Cisco Unified Presence Server.
- C. Do nothing; the "BLF for Call Lists" is enabled by default.
- D. Enable the "BLF for Call Lists" in the Enterprise parameter section.

**Answer:** ([SHOW ANSWER](#))

#### NEW QUESTION: 15

Refer to the exhibit.



CSS1 is assigned to the IP phone that Bob uses. Which matched route pattern allows Bob to place calls to

3330?

- A. Partition D, 3333
- B. Partition B, 3XXX
- C. Partition A, 3XXX
- D. Partition C, 3333

**Answer:** ([SHOW ANSWER](#))

### NEW QUESTION: 16

Which statement regarding Cisco IP voice media streaming application is correct?

- A. It should be activated on the gatekeeper in cluster that supports the TFTP service.
- B. It should be activated on the node in cluster that supports the TFTP service.
- C. It should be activated on the node in cluster that does not support the TFTP service.
- D. It should be activated on the gateway in cluster that supports the TFTP service.

Answer: ([SHOW ANSWER](#))

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### NEW QUESTION: 17

How are Cisco Unified CallManager location parameters used?

- A. Implement call admission control in a centralized call processing deployment.
- B. Specify the bandwidth used for audio and video calls.
- C. Provide alternate call routing when the primary call path is unavailable.
- D. Assign directory numbers to devices as they connect to the IP telephony network.

Answer: ([SHOW ANSWER](#))

### NEW QUESTION: 18

Which two are enterprise parameters? (Choose two.)

- A. Enable Dependency Records
- B. Enable Call Detail Records
- C. T302 parameter
- D. User search limit
- E. codecs parameters for voice media-streaming applications

Answer: ([SHOW ANSWER](#))

### NEW QUESTION: 19

An engineer configures a router with an ISDN PRI connection that only contains two dial peers as follows:

dial-peer voice 10 pots

destination-pattern 9T

port 0/0/0:23

dial-peer voice 100 voip

destination-pattern 45..

session target ipv4:10.2.10.1

When an inbound call is placed from the PSTN to extension 4510, a dial tone is heard. Which additional dial peer is needed to allow the call to terminate directly to the internal number?

**A.** dial-peer voice 1 pots

answer-address .

direct-inward-dial

**B.** dial-peer voice 1 pots

destination-pattern 45..

direct-inward-dial

**C.** dial-peer voice 1 pots

destination-pattern 45..

port 0/0/0:23

direct-inward-dial

**D.** dial-peer voice 1 pots

incoming called-number .

direct-inward-dial

**Answer:** ([SHOW ANSWER](#))

#### NEW QUESTION: 20

Which two commands set up the CoS to DSCP maps on Cisco IOS switches? (Choose two.)

**A.** mls qos

**B.** auto qos global

**C.** qls mls voip auto

**D.** auto qos voip cisco-phone

**E.** qos mls

**Answer:** ([SHOW ANSWER](#))

#### NEW QUESTION: 21

If you have configured translation rules, what command do you use to verify digit manipulation?

**A.** Set translation-rule

**B.** Show translation-rule

**C.** Test translation-rule

**D.** Debug translation-rule

**E.** Execute translation-rule

**Answer:** ([SHOW ANSWER](#))

#### NEW QUESTION: 22

What is the maximum number of 1080p30 HD Conference Participants if an MSE 8000 has four MSE8710 blades clustered?

**A.** 800

- B. 48
- C. 720
- D. 180

Answer: ([SHOW ANSWER](#))

#### NEW QUESTION: 23

Which three steps are needed to enable time-of-day call routing? (Choose three.)

- A. Configure an access list.
- B. Configure a time period.
- C. Configure a time schedule.
- D. Configure a route pattern.
- E. Configure a partition.
- F. Configure dial rules.
- G. Configure a calling search space.

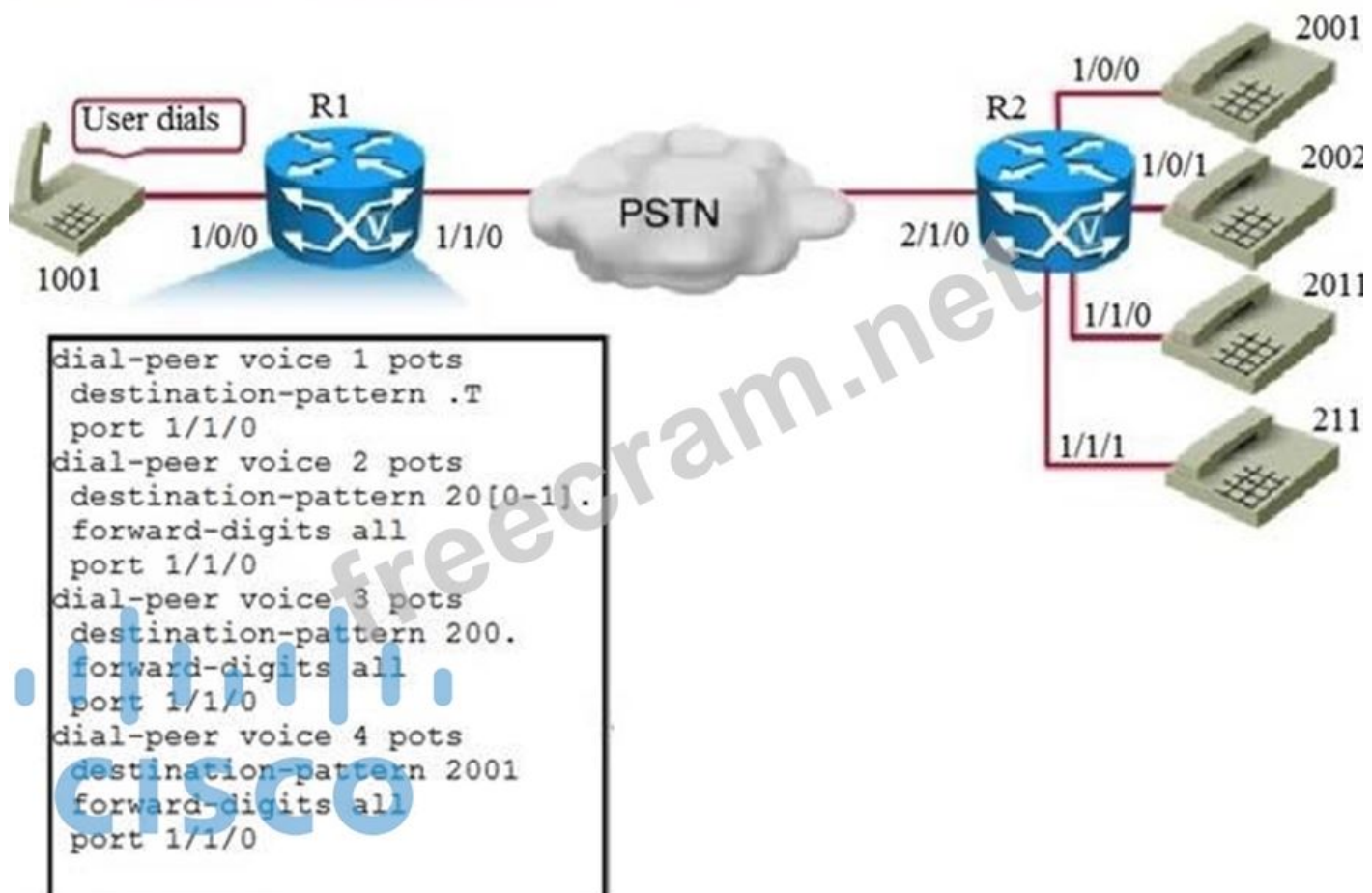
Answer: ([SHOW ANSWER](#))

Explanation/Reference:

Reference: <http://www.cisco.com/c/en/us/support/docs/voice-unified-communications/unified-communications-manager-callmanager/109588-tod-routing-example00.html>

#### NEW QUESTION: 24

Refer to Exhibit:



If the user at extension 1001 dials the user at extension 2111, which dial peer is used to route the call?

- A. 3
- B. 2
- C. 4
- D. 1

Answer: ([SHOW ANSWER](#))

#### NEW QUESTION: 25

Which three of the following are types of video conference? (Choose three.)

- A. permissive
- B. rich media
- C. immersive
- D. rendezvous
- E. scheduled
- F. Ad Hoc

Answer: ([SHOW ANSWER](#))

#### NEW QUESTION: 26

You are a Voice Engineer working at ABC Company. You are asked to configure Cisco Unified Border Element in Cisco Unified Communications Manager. Which two configurations are required? (Choose two.)

- A. Configure a SIP profile with Early Offer support for voice and video calls; insert MTP as needed.
- B. Bind the SIP signaling to the Cisco Unified Border Element interface.
- C. Configure dial peers on Cisco Unified Border Element for inbound and outbound calls appropriately.
- D. Change the SIP trunk security profile to a nonsecure SIP trunk profile.
- E. Under the SIP information configuration, set the destination port to 5061.

Answer: ([SHOW ANSWER](#))

#### NEW QUESTION: 27

When implementing a new dial plan, which tool is used to determine if the dial plan theoretically functions as desired?

- A. CDR
- B. RTMT
- C. DNA
- D. Cisco Prime Collaboration

Answer: ([SHOW ANSWER](#))

#### NEW QUESTION: 28

Which three service types support call classification (on-net/off-net) configurations? (Choose three.)

- A. route pattern
- B. SIP route patterns

- C. route group
- D. gateways devices (H.323, MGCP)
- E. CTI route point
- F. SIP trunk

**Answer:** ([SHOW ANSWER](#))

#### NEW QUESTION: 29

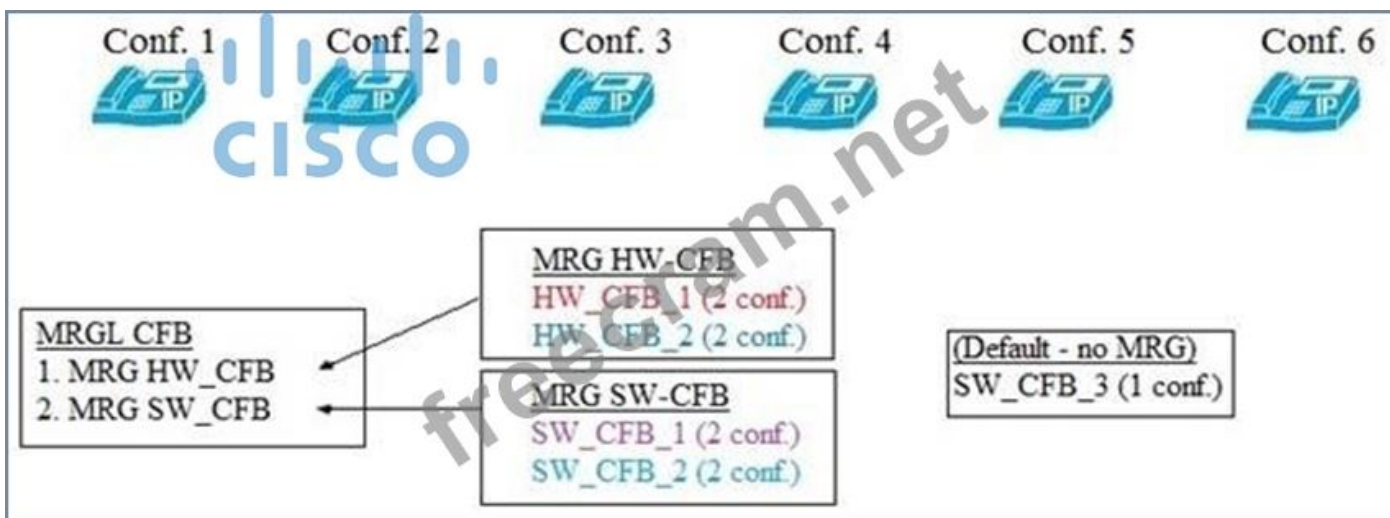
When a user dials a telephone number from a Type A SIP phone, how are the digits analyzed by Cisco Unified Communications Manager if no SIP dial rules are configured?

- A. If there are no dial rules present and the user uses the Dial softkey the digits are sent to the Cisco Unified Communications Manager digit-by-digit.
- B. When no dial rules are present, digits are sent to the Cisco Unified Communications Manager digit-by-digit.
- C. When no dial rules are present, digits are sent to the Cisco Unified Communications Manager en bloc.
- D. With a Type A SIP phone the digits are only sent en bloc.

**Answer:** ([SHOW ANSWER](#))

#### NEW QUESTION: 30

Refer to the exhibit.



Which conference resource is Conference 2 using assuming the MRGL has just been reset and all 6 conferences have become active?

- A. SW\_CFB\_3
- B. SW\_CFB\_1
- C. HW\_CFB\_1
- D. SW\_CFB\_2
- E. HW\_CFB\_2

**Answer:** E ([LEAVE A REPLY](#))

#### NEW QUESTION: 31

What is the purpose of using device pools?

- A. To provide all the IP phone parameters required for auto-registration
- B. To provide an easily modified set of system parameters for the Cisco Unified CallManager server and endpoints
- C. To consolidate the dial plan for a set of IP phones in the same location
- D. To provide a convenient way to define a set of common characteristics that can be assigned to devices

**Answer:** ([SHOW ANSWER](#))

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#### **NEW QUESTION: 32**

Which three Cisco Unified CallManager configuration steps are required to support third party SIP phones?

(Choose three.)

- A. Add the MAC address of the Cisco Unified CallManager server to the SIP phone configuration page
- B. Configure the end user in Cisco Unified CallManager
- C. Configure the phone with the TLS username and password
- D. Associate the device with the end user
- E. Change the proxy address in the SIP phone to an IP address or the Fully Qualified Domain Name (FQDN) of Cisco Unified CallManager
- F. Configure the device in Cisco Unified CallManager

**Answer:** B,D,F ([LEAVE A REPLY](#))

#### **NEW QUESTION: 33**

What function does the device pool perform?

- A. It separates user and physical location information.
- B. It sets the default device parameters for all IP phones in the cluster.
- C. It sets the location bandwidth used between different locations.
- D. It provides a single point of configuration for every device in a Cisco Unified Communications Manager cluster.

**Answer:** A ([LEAVE A REPLY](#))

Explanation/Reference:

Reference: [http://www.cisco.com/c/en/us/td/docs/voice\\_ip\\_comm/cucm/admin/7\\_1\\_2/ccmcfg/bccm-712-cm/b02devpl.html](http://www.cisco.com/c/en/us/td/docs/voice_ip_comm/cucm/admin/7_1_2/ccmcfg/bccm-712-cm/b02devpl.html)

#### NEW QUESTION: 34

Which discard digits instruction removes all digits prior to the North American Numbering Plan portion of the route pattern?

- A. PreDot
- B. Pre
- C. PreNANP
- D. PreAt

Answer: ([SHOW ANSWER](#))

#### NEW QUESTION: 35

Which feature allows TelePresence endpoints to have a multipoint video conference without using MCU resources?

- A. Media Termination Point
- B. Conference Bridge
- C. MultiPoint
- D. MultiSite

Answer: ([SHOW ANSWER](#))

#### NEW QUESTION: 36

Refer to the exhibit.



An engineer is working on a ticket that reports TEL1 cannot call TEL2. These dial peers are configured to route the call between the endpoints.

```
dial-peer voice 2001 voip
destination-pattern 2001
session protocol sipv2
session target ipv4:10.1.1.2
incoming called-number .
codec g722-64
video codec h264
dtmf-relay rtp-nte sip-notify sip-kpml
no vad
dial-peer voice 3001 voip
destination-pattern 3001
session protocol sipv2
```

```
session target ipv4:10.3.1.2
incoming called-number .
codec g722-64
video codec h264
dtmf-relay rtp-nte sip-notify sip-kpml
no vad
```

While troubleshooting, the engineer discovers that the Cisco Unified Border Element is giving the error "SIP/2.0 403 Forbidden". Which two Cisco Unified Border Element configurations resolve this issue? (Choose two.)

**A.** voice service voip

```
mode border-element
```

```
ip address trusted list
```

```
ipv4 10.1.1.0 255.255.255.0
```

```
ipv4 10.2.1.0 255.255.255.0
```

```
allow-connections sip to sip
```

```
sip
```

```
bind control source-interface loopback0
```

```
bind media source-interface loopback0
```

**B.** voice service voip

```
no ip address trusted authenticate
```

```
mode border-element
```

```
allow-connections sip to sip
```

```
sip
```

```
bind control source-interface loopback0
```

```
bind media source-interface loopback0
```

**C.** voice service voip

```
mode border-element
```

```
ip address trusted list
```

```
ipv4 10.1.1.0 255.255 255.0
```

```
ipv4 10.3.1.0 255.255.255.0
```

```
allow-connections sip to sip
```

```
sip
```

```
bind control source-interface loopback0
```

```
bind media source-interface loopback0
```

**D.** voice service voip

```
mode border-element
```

```
ip address trusted list
```

```
allow-connections sip to sip
```

```
sip
```

```
bind control source-interface loopback0
```

```
bind media source-interface loopback0
```

E. voice service voip  
mode border-element  
allow-connections sip to sip  
sip  
bind control source-interface loopback0  
bind media source-interface loopback0

**Answer:** ([SHOW ANSWER](#))

#### **NEW QUESTION: 37**

Which two options are components of Media Resources Architecture? (Choose two.)

- A. Music on hold
- B. Media Resource Manager
- C. Media Resource Group
- D. Route list
- E. Route group

**Answer:** ([SHOW ANSWER](#))

Explanation/Reference:

Reference: [https://www.cisco.com/c/en/us/td/docs/voice\\_ip\\_comm/cucm/srnd/collab09/clb09/media.html#60693](https://www.cisco.com/c/en/us/td/docs/voice_ip_comm/cucm/srnd/collab09/clb09/media.html#60693)

#### **NEW QUESTION: 38**

Which tool can you use to verify and troubleshoot a dial plan?

- A. Audit Log Configuration
- B. Dialed Number Analyzer
- C. CDR Analysis and Reporting
- D. Troubleshooting Trace Settings

**Answer:** ([SHOW ANSWER](#))

#### **NEW QUESTION: 39**

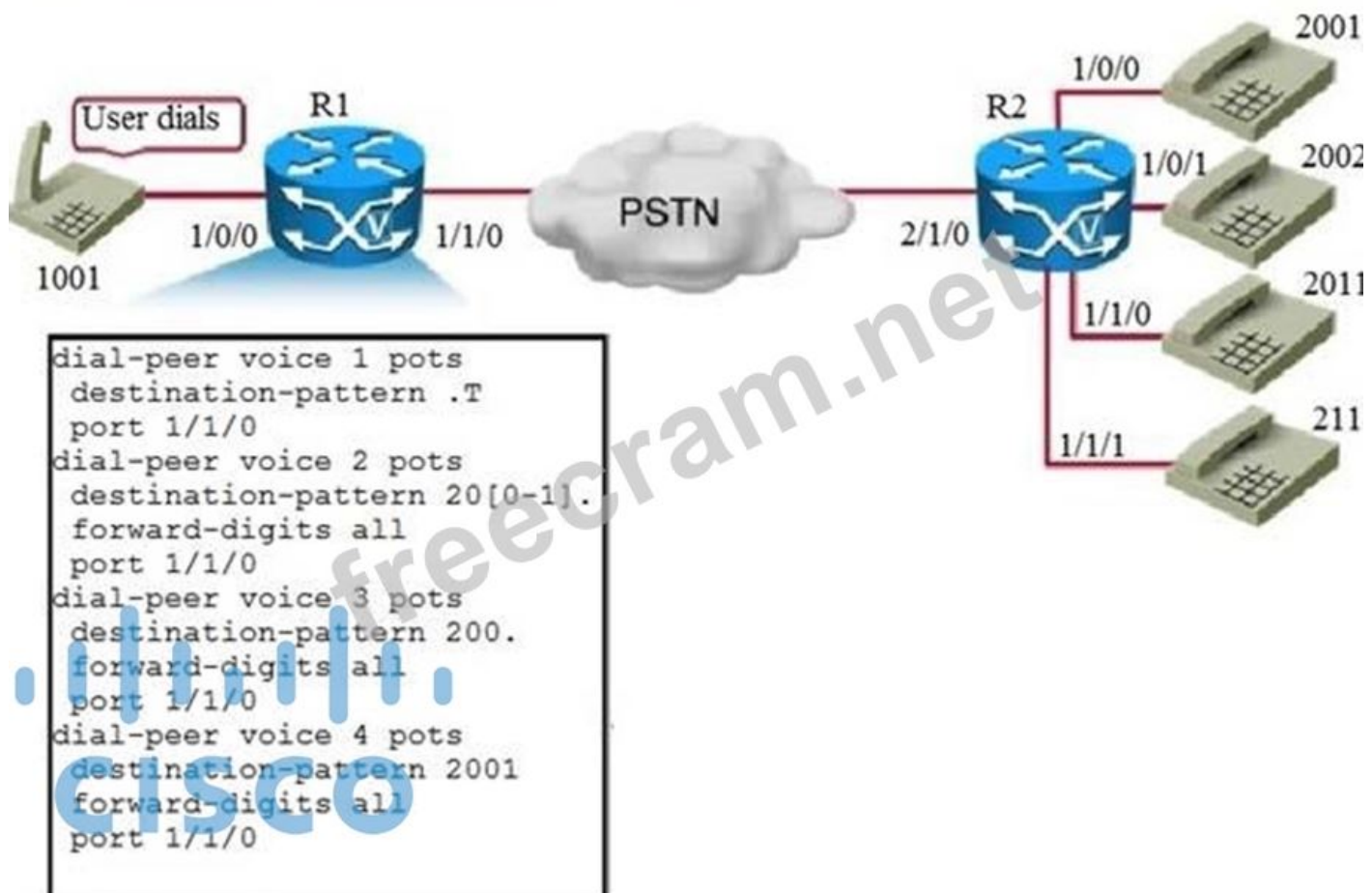
Which two describe how media resources and audio streams operate with Cisco Unified Communications Manager? (Choose two.)

- A. Audio streams are always terminated by media resources.
- B. Signaling between Cisco Unified Communications Manager and hardware media resources uses the same protocol as the call-signaling type.
- C. Audio streams are terminated differently depending on the signaling protocol, such as SCCP or SIP.
- D. All media resources register with Cisco Unified Communications Manager.
- E. There are no situations in which IP phone-to-IP phone audio streams will go directly between endpoints.

**Answer:** ([SHOW ANSWER](#))

#### **NEW QUESTION: 40**

Refer to Exhibit:



If the user at extension 1001 dials the user at extension 2001, which dial peer is used to route the call?

- A. 2
- B. 4
- C. 1
- D. 3

Answer: ([SHOW ANSWER](#))

#### NEW QUESTION: 41

What are the three characteristics that govern the type of Cisco Unified Communications Manager deployment model? (Choose three.)

- A. Geographical distribution
- B. Services provided by the PSTN carrier
- C. Type of applications
- D. Size
- E. Network characteristics
- F. Number of applications

Answer: ([SHOW ANSWER](#))

#### NEW QUESTION: 42

When CUCM is installed, how can the Customer Use the Operating system?

- A. GUI
- B. CLI
- C. none
- D. GUI and CLI
- E. OS can be accessed by CISCO TAC only

Answer: ([SHOW ANSWER](#))

#### NEW QUESTION: 43

Which action does the urgent priority flag take when set on a route pattern?

- A. It makes the user enter a FAC code.
- B. It waits for inter-digit timeout.
- C. It routes the call immediately.
- D. It waits for all digits to be collected.

Answer: ([SHOW ANSWER](#))

#### NEW QUESTION: 44

Refer to the exhibit.

```
Voice-HUB#show isdn status
Global ISDN Switchtype - primary-dms100
ISDN Serial2/0/0:23 interface
dsl 0, interface ISDN Switchtype - primary-dms100
Layer 1 Status:
ACTIVE
Layer 2 Status:
TEI - 0, Ces - 1, SAPI - 0, State - MULTIPLE_FRAME_ESTA
Layer 3 Status:
0 Active Layer 3 Call(s)
Active dsl 0 CCBs - 0
The Free Channel Mask: 0x807FFFFF
Number of L2 Discards - 0, L2 Session ID - 4
Total Allocated ISDN CCBs - 0
```

An engineer is trying to determine the status of the ISDN D Channel on a new PRI circuit, types in the show isdn status command, and sees this output. Which two pieces of information can be determined from this output? (Choose two.)

- A. No active calls are established.
- B. The provider is using the 5ESS protocol.
- C. The D channel is working correctly.
- D. The connection to the Cisco Unified Communications Manager has been established.
- E. There are four B channels.

Answer: ([SHOW ANSWER](#))

#### NEW QUESTION: 45

Employees at a company must dial 9 first to reach an external number. Which route pattern is used to route the call if a user dials the toll-free number 1-866-555-1212?

- A. 9.18XX[2-9]XXXXXX
- B. 9.1[2-9]XX[2-9]XXXXXX
- C. 9.1XXXXXXXXXX
- D. 9.18[0,5-8][0,5-8]XXXXXX
- E. 9.1866[2-9]XXXXXX
- F. 9.1866XXXXXX

Answer: ([SHOW ANSWER](#))

#### NEW QUESTION: 46

Which statement about the relationship between gateways and route groups is true?

- A. Each gateway can belong to multiple route groups and can be listed only once within each route group.
- B. Each gateway can belong to only one route group and can be listed only once within that route group.
- C. Each gateway can belong to only one route group and can be listed multiple times within that route group.
- D. Each gateway can belong to multiple route groups and can be listed multiple times within each route group.

Answer: ([SHOW ANSWER](#))

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#### NEW QUESTION: 47

Which action must you perform before you update Cisco Unified Communications Manager using BAT?

- A. Enable promiscuous mode on the Cisco Unified CM BAT database.
- B. Create or edit a CMC or FAC CSV file and insert the file in BAT.
- C. Put Cisco Unified CM in BAT service mode.
- D. Temporarily disconnect all non-BAT related servers, such as Cisco Presence.

Answer: ([SHOW ANSWER](#))

#### NEW QUESTION: 48

When an IP phone boots up, which mechanism provides the VLAN ID?

- A. Option 150

- B. DHCP
- C. CDP
- D. TFTP
- E. DNS

**Answer:** ([SHOW ANSWER](#))

#### NEW QUESTION: 49

Which two of the following are types of video conference? (Choose two.)

- A. rendezvous
- B. rich media
- C. scheduled
- D. immersive
- E. permissive

**Answer:** ([SHOW ANSWER](#))

Explanation/Reference:

Reference:

[https://www.cisco.com/c/en/us/td/docs/voice\\_ip\\_comm/cucm/srnd/collab09/clb09/confernc.pdf](https://www.cisco.com/c/en/us/td/docs/voice_ip_comm/cucm/srnd/collab09/clb09/confernc.pdf) page 11-2 and 11-3.

#### NEW QUESTION: 50

Managers have asked to be able to have an extension call them back if it is busy when they try to call it. In what two places can this feature be assigned to the Managers phone? (Choose two.)

- A. Device Settings > Phone Button Template
- B. Device Settings > Softkey Template
- C. Device Settings > Firmware Load Information
- D. Device Settings > Device Defaults
- E. Device Settings > Device Profile
- F. Device Settings > Default Device Profile Configuration

**Answer:** ([SHOW ANSWER](#))

#### NEW QUESTION: 51

Which two of these are reasons to deploy media resource access control in a Cisco Unified Communications Manager cluster? (Choose two.)

- A. to enable Cisco Unified Communications Manager to share and access the resources that is available across multiple clusters.
- B. to allow media resource access control based on the type of the resource.
- C. to enable media resource access control based on least used resource.
- D. to enable Cisco Unified Communications Manager to perform load distribution within a group of dissimilar media resources.
- E. to enable hardware and software media resources to coexist within a Cisco Unified Communications Manager and to be used with different priorities.

**Answer: (SHOW ANSWER)**

**NEW QUESTION: 52**

What is the maximum number of call processing servers supported in a single cluster?

- A. 24
- B. 12
- C. 10
- D. 5
- E. 20
- F. 8

**Answer: (SHOW ANSWER)**

**NEW QUESTION: 53**

To verify that proper digit manipulation is being applied when using called party transformation patterns, which Cisco IOS command do you use from the router or voice gateway?

- A. show voice ccapi inout
- B. debug voice ccapi in
- C. show voice ccapi
- D. show voice ccapi in
- E. debug voice ccapi inout
- F. debug voice ccapi out
- G. show voice ccapi out

**Answer: (SHOW ANSWER)**

Explanation/Reference:

Reference: <http://www.netcraftsmen.com/called-party-transformation-patterns-example-and-validation/>

**NEW QUESTION: 54**

Which digit manipulation command works under a VoIP dial peer?

- A. forward-digits
- B. digit-strip
- C. prefix
- D. translation-profile

**Answer: (SHOW ANSWER)**

**NEW QUESTION: 55**

Which two codecs are supported by the Cisco Unified Communications Manager software conferencing media resource? (Choose two.)

- A. wideband
- B. G.723
- C. GSM
- D. G.711

E. G.729

Answer: ([SHOW ANSWER](#))

**NEW QUESTION: 56**

Which two Cisco multiparty bridges are supported when you set up WebEx-enabled Cisco TelePresence?

(Choose two.)

- A. Cisco TelePresence Server
- B. Cisco Unified Communications Manager SME
- C. Cisco IOS gateways
- D. Cisco TelePresence MCU
- E. Cisco Unified Border Element

Answer: ([SHOW ANSWER](#))

**NEW QUESTION: 57**

What are two characteristics of a Cisco Unified Communications Manager dial plan? (Choose two.)

- A. partitions and calling search spaces
- B. COR
- C. dial peers
- D. voice translation rules
- E. route patterns

Answer: A,E ([LEAVE A REPLY](#))

Explanation/Reference:

Reference: [http://www.cisco.com/c/en/us/td/docs/voice\\_ip\\_comm/cucm/admin/8\\_6\\_1/ccmcfg/bccm-861-cm/b03rtpat.html](http://www.cisco.com/c/en/us/td/docs/voice_ip_comm/cucm/admin/8_6_1/ccmcfg/bccm-861-cm/b03rtpat.html)

**NEW QUESTION: 58**

Refer to the exhibit.



```
911
9.[2-9]XXXXXX
9.1[2-9]XX[2-9]XXXXXX
9.18[0,6-8][0,6-8]XXXXXXX
9.1900XXXXXXX
9011.!
```

To reach an external number, a user first dials a 9. Which route pattern will be matched if a user dials the toll-free number 18005550123?

- A. 9.18[0,6-8][0,6-8]XXXXXXX
- B. 9.1[2-9]XX[2-9]XXXXXX
- C. 9.[2-9]XXXXXX

D. 9.1900XXXXXXX

E. 9011!

F. 911

**Answer:** ([SHOW ANSWER](#))

#### NEW QUESTION: 59

Which route pattern is used when 25911 is dialed from an endpoint with a CSS that can access the required partition?

A. 259XX

B. 2[56]XXX

C. 25XXX

D. 2591[12]

**Answer:** ([SHOW ANSWER](#))

#### NEW QUESTION: 60

Which groups will have access to the Meet-Me conference numbers that are set to use the <None> partition?

A. All users will have access to the Meet-Me conference numbers.

B. Any IP phones with their lines configured in the <None> partition will have access to the Meet-Me conference numbers.

C. Only users that have the <None> partition assigned in their calling search space will have access to the Meet-Me conference numbers.

D. The <None> partition will prevent all users from accessing the Meet-Me conference numbers.

**Answer:** ([SHOW ANSWER](#))

#### NEW QUESTION: 61

Which three devices are analyzed by the Cisco Unified Communications Dialed Number Analyzer?  
(Choose three.)

A. gatekeeper

B. remote destination

C. CTI route point

D. trunk

E. phone

F. gateway

**Answer:** D,E,F ([LEAVE A REPLY](#))

Explanation/Reference:

Reference: [http://www.cisco.com/c/en/us/td/docs/voice\\_ip\\_comm/cucm/dna/8\\_6\\_1/dnaguide/dnaan.html#wp1227993](http://www.cisco.com/c/en/us/td/docs/voice_ip_comm/cucm/dna/8_6_1/dnaguide/dnaan.html#wp1227993)

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**NEW QUESTION: 62**

Which three characteristics are used to determine which devices go into a device pool? (Choose three.)

- A. Geographic proximity
- B. User hold MOH source
- C. Class of service
- D. Extension mobility CSS
- E. auto-registration CSS
- F. Device type

Answer: ([SHOW ANSWER](#))

**NEW QUESTION: 63**

Which two conferencing endpoints support the MultiSite video conferencing feature? (Choose two.)

- A. SX10
- B. SX90
- C. SX60
- D. SX20
- E. SX80

Answer: ([SHOW ANSWER](#))

**NEW QUESTION: 64**

Which two features are advantages of deploying a cluster over WAN? (Choose two.)

- A. Shared line appearances
- B. Centralized DSP resources
- C. Extension mobility within the cluster
- D. Failover to co-resident Cisco Unified Call Manager Express platforms
- E. Scalability up to 20 sites, if there is sufficient IP WAN bandwidth

Answer: ([SHOW ANSWER](#))

**NEW QUESTION: 65**

Which type of IOS Conference bridge requires all users to use the same video format in order to participate in a video conference?

- A. Ad Hoc Conferencing
- B. Homogenous Conferencing

- C. Heterogenous Conferencing
- D. Guaranteed Audio Video Conferencing
- E. Meet-Me Conferencing

**Answer:** ([SHOW ANSWER](#))

Explanation/Reference:

Explanation :- A homogenous conference refers to one in which participants connect to the ISR G2 with devices that support the same video format attributes (for example, the same codec, resolution, frame rate, and bit rate).

#### **NEW QUESTION: 66**

In Cisco Unified Communications and Collaboration systems, generally, the dial plan is responsible for instructing the call processing agent on how to route calls. Which two main functions other than calling privileges, path selection, and endpoint addressing does the dial plan perform? (Choose two.)

- A. trunk priority
- B. presentation of information about identities involved in the call
- C. codec selection
- D. skills preference
- E. manipulation of dialed destination

**Answer:** B,E ([LEAVE A REPLY](#))

Explanation/Reference:

Reference: [https://www.cisco.com/c/en/us/td/docs/voice\\_ip\\_comm/cucm/srnd/collab09/clb09/dialplan.pdf](https://www.cisco.com/c/en/us/td/docs/voice_ip_comm/cucm/srnd/collab09/clb09/dialplan.pdf)

#### **NEW QUESTION: 67**

What is the correct order for configuring call routing in Cisco Unified Communications Manager?

- A. gateway and SIP trunk, route list, route group, route pattern
- B. gateway and SIP trunk, route group, route list, route pattern
- C. gateway and SIP trunk, route group, route pattern, route list
- D. route list, gateway and SIP trunk, route group, route pattern

**Answer:** ([SHOW ANSWER](#))

#### **NEW QUESTION: 68**

When dialing off-hook from an SCCP phone, how does Cisco Unified Communications Manager analyze the incoming dialed digits?

- A. digit-by-digit analysis
- B. SIP dial rules
- C. KPML
- D. en bloc dialing

**Answer:** ([SHOW ANSWER](#))

#### **NEW QUESTION: 69**

Which three of these are characteristics of a Cisco Unified Communications Manager software- based Media Termination Point? (Choose three.)

- A. The codec and packetization of both call legs must be identical.
- B. This Media Termination Point type does not require any DSP resources on the Cisco router.
- C. It can provide G.711 mu-law to G.711 a-law conversion and vice versa.
- D. This Media Termination Point type is typically used for RSVP agent configurations or Cisco Unified Border Element media flow-through configurations.
- E. It can provide packetization conversion for a given codec (for example, if one call leg is using a 20-ms sample size, but the other call leg is using a 30-ms sample size).
- F. DSP resources are required.

**Answer:** ([SHOW ANSWER](#))

#### NEW QUESTION: 70

Which tool can you use to evaluate dial plans for errors and to examine internal-to-internal and internal-to- external calls?

- A. RTMT
- B. DNA
- C. QRT Viewer
- D. CAR tool

**Answer:** ([SHOW ANSWER](#))

Explanation/Reference:

Reference: [http://www.cisco.com/c/en/us/td/docs/voice\\_ip\\_comm/cucm/dna/9\\_1\\_1/](http://www.cisco.com/c/en/us/td/docs/voice_ip_comm/cucm/dna/9_1_1/CUCM_BK_C514A4AE_00_cucm-dialed-number-analyzer-91/CUCM_BK_C514A4AE_00_cucm-dialed-number-analyzer-91_chapter_01.html)

CUCM\_BK\_C514A4AE\_00\_cucm-dialed-number-analyzer-91/CUCM\_BK\_C514A4AE\_00\_cucm-dialed-number-analyzer-91\_chapter\_01.html

#### NEW QUESTION: 71

Which two call coverage features are used by voice gateways? (Choose two.)

- A. call Pickup
- B. call hunt
- C. overlaid phone-dn
- D. call redirecting
- E. call Park

**Answer:** ([SHOW ANSWER](#))

Explanation/Reference:

#### NEW QUESTION: 72

Which two SIP issues should be considered when deploying KPML? (Choose two.)

- A. Additional MTP resources will be required.
- B. There will be additional processing demands on the Cisco Unified Communications Manager server.
- C. There will be significantly more signaling traffic on the network.
- D. Social engineering will be required to educate users about how to dial a call with SIP.

E. Additional configuration steps will be required for SIP phone dial rules.

**Answer:** ([SHOW ANSWER](#))

#### NEW QUESTION: 73

Users that are subscribed to the default corporate directory are complaining that when they press the Directories button, they receive an error message regarding the URL. Which two parameters need to be edited? (Choose two.)

- A. Select the Service Parameter for the appropriate server and Cisco CallManager (active) and change the URLs for each service to match the DNS name or IP address of the server being configured.
- B. Verify that the corporate directory has the Enabled check box checked in the Enterprise Parameter > Phone URL Parameters page.
- C. Reset the Enterprise Phone URL Parameters.
- D. Verify that the corporate directory has the Enabled check box checked in the IP Phone Services > Configuration Service Information page.
- E. Reset the Service Parameter > Phone URL Parameters.
- F. Change the URLs under Enterprise Parameters to match the DNS name or the IP address being used by the server being configured and reset the IP phones.

**Answer:** D,F ([LEAVE A REPLY](#))

#### NEW QUESTION: 74

Which two statements describe the function of the off-net option in a route pattern? (Choose two.)

- A. This option could be used to block an external call from transferring to an external party.
- B. This option could be used to block an external call from transferring to an internal party.
- C. The off-net option indicates that if a call is not routed through this route pattern, it is considered off the local network, or "off-net".
- D. The off-net option indicates that if a call is routed through this route pattern, it is considered off the local network, or "off-net."

**Answer:** ([SHOW ANSWER](#))

#### NEW QUESTION: 75

An engineer is setting up a system with voice and video endpoints using auto-QoS policy on the switches.

Which DSCP values are expected for interactive voice and video?

- A. EF and AF41
- B. EF and CS6
- C. EF and CS3
- D. EF and AF21

**Answer:** ([SHOW ANSWER](#))

Explanation/Reference:

Reference:

[http://www.cisco.com/c/en/us/td/docs/solutions/Enterprise/WAN\\_and\\_MAN/QoS\\_SRND/QoS-SRND-Book/QoSIntro.html](http://www.cisco.com/c/en/us/td/docs/solutions/Enterprise/WAN_and_MAN/QoS_SRND/QoS-SRND-Book/QoSIntro.html)

#### NEW QUESTION: 76

What happens when the TFTP server is not available?

- A. CUCM can't read phone configurations
- B. TFTP server needs to be restarted
- C. Phones need to be restarted
- D. Phones can't download configuration

Answer: ([SHOW ANSWER](#))

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#### NEW QUESTION: 77

Which statement about the DiffServ QoS deployment architecture is true?

- A. DiffServ architecture provides QoS by assuring treatment for a specific traffic flow.
- B. DiffServ architecture differentiates and classifies various types of traffic and provides several levels of service based on that classification.
- C. DiffServ does not differentiate traffic types and thus cannot provide service levels.
- D. Resource Reservation Protocol is an example of a DiffServ mechanism where each router on the path for packet transmission is informed of the upcoming packet stream.

Answer: ([SHOW ANSWER](#))

#### NEW QUESTION: 78

Which four software based media resources require that the Cisco IP voice media stream Application be activated?

- A. MOH
- B. SIP
- C. Audio conferencing
- D. MTP
- E. Annunciator
- F. Gatekeeper
- G. H 323 Gateways

**Answer: A,C,D,E ([LEAVE A REPLY](#))**

**NEW QUESTION: 79**

Which two options are functions of Cisco Unified Communications Manager? (Choose two.)

- A. Call processing
- B. Transcoding
- C. Directory services
- D. Rich-media conferencing

**Answer: ([SHOW ANSWER](#))**

**NEW QUESTION: 80**

Which three events describe the relationship of an IP phone to its secondary subscriber in a Cisco Unified Communications Manager system? (Choose three.)

- A. The IP Phone sends a TCP keepalive message every 60 seconds.
- B. The IP Phone registers with all the Cisco Unified Communications Manager subscribers in the group simultaneously, including the secondary subscriber.
- C. The IP Phone sends a TCP keepalive message every 30 seconds.
- D. The IP Phone registers with the primary system when the secondary system is unavailable.
- E. The IP Phone attempts to register with the secondary system when the primary system is unavailable.
- F. The IP Phone sends a TCP connect message.

**Answer: ([SHOW ANSWER](#))**

**NEW QUESTION: 81**

Which traffic-conditioning mechanism buffers packets during congestion, instead of dropping them?

- A. shaping
- B. policing
- C. class-selector PHB
- D. packet fragmentation and interleaving
- E. PHB

**Answer: ([SHOW ANSWER](#))**

**NEW QUESTION: 82**

Which two dial peers successfully forward the digits 5551212? (Choose two.)

- A. dial-peer voice 1 pots  
destination-pattern 1408.....  
port 1/1
- B. dial-peer voice 1 pots  
destination-pattern 1408.....  
forward-digits all  
port 1/1
- C. dial-peer voice 1 pots

destination-pattern 555....

no digit-strip

port 1/1

**D.** dial-peer voice 1 pots

destination-pattern 14085551212

forward-digits 10

port 1/1

**E.** dial-peer voice 1 pots

destination-pattern 14085551...

prefix 555

port 1/1

**Answer:** ([SHOW ANSWER](#))

### NEW QUESTION: 83

Which two CLI commands allow you to verify the correct VoIP dial-peer matching? (Choose two.)

**A.** debug voip dialpeer inout

**B.** debug ccsip messages

**C.** show dialplan number

**D.** debug dialpeer match

**E.** debug voip ccapi service

**Answer:** ([SHOW ANSWER](#))

### NEW QUESTION: 84

Which route pattern is the best match for the range of 6000 to 6599?

**A.** 6[05]XX

**B.** 6[0-5]XX

**C.** 6XXX

**D.** 65XX

**E.** 60XX

**Answer:** ([SHOW ANSWER](#))

### NEW QUESTION: 85

Which three Cisco Unified Communications Manager settings can be used to support IP phone services?

(Choose three.)

**A.** Service Parameters

**B.** Phone Configuration window

**C.** IP Phone Service Configuration window

**D.** Enterprise Parameters

**E.** Device Defaults

**F.** Common Phone Profile

**Answer:** ([SHOW ANSWER](#))

**NEW QUESTION: 86**

Which statement is true?

- A. VoIP dial peers use the SIP signaling protocol by default.
- B. VoIP dial peers are POTS by default.
- C. VoIP dial peers use the MGCP signaling protocol by default.
- D. VoIP dial peers use the H.323 signaling protocol by default.

**Answer:** ([SHOW ANSWER](#))

**NEW QUESTION: 87**

Which two of these are characteristics of a Cisco Unified Communications Manager software conference resource? (Choose two.)

- A. It supports only unicast audio streams.
- B. If Cisco Unified Communications Manager is co-resident on the same server, the maximum number of audio streams per server is 46.
- C. It supports both unicast and multicast audio streams.
- D. Any combination of G.711 mu-law, G.711 a-law, or wideband audio streams may be connected.
- E. The number of participants is based on the number of DSP resources available.
- F. It supports any combination of codec types.

**Answer:** ([SHOW ANSWER](#))

**NEW QUESTION: 88**

An engineer is configuring QoS for a Cisco Unified Communications Manager implementation. Which two markings can be used for voice traffic? (Choose two.)

- A. cos 4
- B. ip precedence 3
- C. ip precedence 5
- D. cos 6
- E. dscp ef
- F. dscp 5

**Answer:** ([SHOW ANSWER](#))

**NEW QUESTION: 89**

Which pattern does the route pattern \+61278[^3-5]59XXX match?

- A. +61278259001
- B. +61278459001
- C. \+61278359001
- D. 278259001
- E. +612783559001

**Answer:** ([SHOW ANSWER](#))

### NEW QUESTION: 90

Which three options can belong to specific partitions? (Choose three.)

- A. voice-mail servers
- B. directory numbers
- C. translation patterns
- D. gateways
- E. IP phones
- F. route patterns

**Answer: B,C,F** ([LEAVE A REPLY](#))

Explanation/Reference:

Reference: [http://www.cisco.com/c/en/us/td/docs/voice\\_ip\\_comm/cucm/admin/7\\_1\\_2/ccmcfg/bccm-712-cm/b02devpl.html](http://www.cisco.com/c/en/us/td/docs/voice_ip_comm/cucm/admin/7_1_2/ccmcfg/bccm-712-cm/b02devpl.html) (configuring a device pool)

### NEW QUESTION: 91

When using video in the Cisco Unified Border Element, which RTP flow offers no protection from the address hiding perspective?

- A. delayed offer
- B. flow-around
- C. early offer
- D. flow-through

**Answer: (**[SHOW ANSWER](#)**)**

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### NEW QUESTION: 92

When a Cisco Unified Communications Manager group has its primary, secondary, and tertiary servers changed, how are those changes propagated to the IP phones registered to the group?

- A. The IP phones will automatically register with the new primary server.
- B. The primary Cisco Unified Communications Manager server automatically pushes a new configuration file to the IP phones in the affected device pool.
- C. The IP phones will need to be reset so that a new configuration file can be downloaded from the TFTP server.

**D.** The device pools will need to be edited so that the new Cisco Unified Communications Manager group information is applied to the appropriate device pools.

**Answer:** ([SHOW ANSWER](#))

#### **NEW QUESTION: 93**

An engineer has added new SIP phones to the Cisco Unified Communications Manager and now can no longer get an outside dial tone when pressing the steering digit 9 for calling local or long distance numbers.

Which action must the engineer take to resolve this issue?

- A.** Configure the outside dial tone of the new SIP phone.
- B.** Check the dial plan to ensure that any new route patterns added in Cisco Unified CM do not start with 9
- C.** Verify that the new phones added in Cisco Unified CM do not start with an extension of 9.
- D.** Ensure the gateway long distance and local dial peers are being triggered on outbound calls.

**Answer:** ([SHOW ANSWER](#))

#### **NEW QUESTION: 94**

Which statement is true about digit analysis within Cisco Unified Communications Manager when all patterns are in the same partition?

- A.** Transformation patterns are always matched first before any other patterns.
- B.** Translation patterns are always matched before route patterns.
- C.** SIP route patterns are always matched before any other patterns.
- D.** Route patterns are always matched before translation patterns.

**Answer:** ([SHOW ANSWER](#))

#### **NEW QUESTION: 95**

Callers inform you that they hear a second dial tone after dialing the number 4085550123. Which dial-peer configuration command resolves this issue?

- A.** forward-digits all
- B.** answer-address 408555....
- C.** destination-pattern 91[2-9]..[2-9].....
- D.** direct-inward-dial

**Answer:** ([SHOW ANSWER](#))

#### **NEW QUESTION: 96**

What are the minimum configuration parameters required to manually add an IP phone to Cisco Unified Communications Manager?

- A.** MAC address, device profile, user security profile, and phone button template
- B.** MAC address, device pool, and phone button template
- C.** MAC address, device pool, phone security profile, and phone button template
- D.** MAC address, user profile, phone security profile, and phone button template

**Answer:** ([SHOW ANSWER](#))

**NEW QUESTION: 97**

When a new subscriber is added to a Cisco Unified Communications Manager cluster, how must the subscriber be configured so that it can receive database updates?

- A.** The subscriber uses the same security password configured in the publisher when it was created.
- B.** The subscriber uses the IDS security password.
- C.** The subscriber uses the subscriber security password configured in the publisher.
- D.** The subscriber uses the security password configured when the subscriber was configured. This is also configured in the publisher and validated on the initial replication of the database.

**Answer:** **A** ([LEAVE A REPLY](#))

**NEW QUESTION: 98**

In the Cisco Unified Communications Manager Administration window, which MOH parameter can be selected for phones and gateways?

- A.** Hold stream volume
- B.** Media resource group list
- C.** Hold stream source
- D.** Media loop timer

**Answer:** ([SHOW ANSWER](#))

**NEW QUESTION: 99**

Which two Cisco TelePresence solutions are available as blades for the Cisco TelePresence MSE 8000 Series chassis? (Choose two.)

- A.** Cisco TelePresence MCU
- B.** Cisco VCS
- C.** Cisco Expressway
- D.** Cisco TelePresence Conductor
- E.** Cisco TelePresence Server

**Answer:** ([SHOW ANSWER](#))

**NEW QUESTION: 100**

You have two groups of managers that need to hold Meet-Me conferences. You want each group to have access to a range of 10 Meet-Me numbers. What are the two best ways to accomplish this? (Choose two.)

- A.** Create two different partitions and add both groups of 10 Meet-Me conference numbers to the partitions.
- B.** Create two different groups of Meet-Me conference numbers, add them to the partition, and add the partition to the two manager groups to provide limited access.
- C.** Ensure that the 10 Meet-Me conference numbers overlap with existing ad hoc conferencing numbers.

**D.** Create a different partition for each group of managers and assign a set of 10 Meet-Me conference numbers to each.

**E.** Create two groups of 10 Meet-Me conference numbers that do not overlap with any other number in the numbering plan.

**Answer:** ([SHOW ANSWER](#))

#### **NEW QUESTION: 101**

Which configuration parameter may need to be adjusted when the Region setting is changed?

**A.** Class of Control

**B.** Device Pool

**C.** Location

**D.** AAR Group

**Answer:** **C** ([LEAVE A REPLY](#))

#### **NEW QUESTION: 102**

Which QoS model is bandwidth-dedicated to provide sufficient quality for applications that need high resolution, such as video terminals?

**A.** DiffServ

**B.** Best-Effort

**C.** IntServ

**D.** DiffServ

**Answer:** ([SHOW ANSWER](#))

#### **NEW QUESTION: 103**

Which of the following simplifies and enhances conference resource management?

**A.** Cisco TelePresence Server

**B.** Cisco TelePresence MCU

**C.** Cisco TelePresence MSE 8000

**D.** Cisco IOS routers with packet voice/data module (PVDM)

**E.** Cisco TelePresence Conductor

**Answer:** ([SHOW ANSWER](#))

Explanation/Reference:

Reference: [http://www.secureitstore.com/datasheets/Collaboration/Cisco-TelePresence-Conductor\\_DS.pdf](http://www.secureitstore.com/datasheets/Collaboration/Cisco-TelePresence-Conductor_DS.pdf)

#### **NEW QUESTION: 104**

Which two of these describe where phone services can appear on an IP phone and who can subscribe to phone services? (Choose two.)

**A.** administrator

**B.** Services and Directories buttons

**C.** Services button

- D. end user and administrator
- E. Services, Directories, and Messages buttons
- F. end user

**Answer:** ([SHOW ANSWER](#))

#### **NEW QUESTION: 105**

By default, what value is voice traffic marked as on a Cisco Catalyst Switch?

- A. DSCP EF value 46, Cos 7
- B. DSCP EF value 46, Cos 5
- C. DSCP CAS6 value 21, Cos 3
- D. DSCP AF21 value 46, Cos 5

**Answer:** ([SHOW ANSWER](#))

#### **NEW QUESTION: 106**

Which of these is a characteristic of the Cisco Unified Communications Manager Bulk Administration Tool in Cisco Unified Communications Manager?

- A. Cisco Unified Communications Manager Auto-Register Phone Tool requires a Customer Response System.
- B. Cisco Unified Communications Manager BAT pages are available from the Serviceability page.
- C. Cisco Unified Communications Manager Auto-Register Phone Tool is part of Bulk Administration.
- D. Cisco Unified Communications Manager BAT needs to be installed from the plug-in page.

**Answer:** ([SHOW ANSWER](#))

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#### **NEW QUESTION: 107**

Which two of the following DSPs can be used in heterogenous conferencing to achieve the required trans- sizing and transcoding functionality? (Choose two.)

- A. PVDM3-256
- B. PVDM3-32
- C. PVDM3-192
- D. PVDM-192
- E. PVDM2-256
- F. PVDM2-128

**Answer: (SHOW ANSWER)**

**NEW QUESTION: 108**

When configuring Cisco Unified Communications Manager, which configuration enables phones to automatically reregister to a Cisco Unified Communications Manager publisher when the connection to the subscriber is lost?

- A. Cisco Unified CM Group
- B. Device Pool
- C. SRST
- D. Route Group

**Answer: (SHOW ANSWER)**

**NEW QUESTION: 109**

Which statement about traffic policing and shaping is true?

- A. Shaping limits traffic rates by dropping, re-marking, or transmitting traffic.
- B. Policing involves regulating excessive traffic rates by delaying (that is, buffering) traffic.
- C. Traffic policing and shaping help regulate bandwidth usage by limiting the amount of traffic.
- D. Traffic policing is more suitable for lower-speed links such as Multilink PPP and Frame Relay, as it buffers excess traffic.

**Answer: (SHOW ANSWER)**

Explanation/Reference:

Reference: [https://books.google.com.pk/books?id=0DOXAwAAQBAJ&pg=PA41&lpg=PA41&dq=Traffic+policing+and+shaping+help+regulate+bandwidth+usage+by+limiting+the+amount+of+traffic&source=bl&ots=YmvtFVj8Op&sig=WrlLCb3w2fEXwSFaBJZnQJ1BCk4&hl=en&sa=X&ved=0ahUKTwig-dOx\\_tbMAhVHRY8KHV4VB9EQ6AEIKDAC#v=onepage&q=Traffic%20policing%20and%20shaping%20help%20regulate%20bandwidth%20usage%20by%20limiting%20the%20amount%20of%20traffic&f=false](https://books.google.com.pk/books?id=0DOXAwAAQBAJ&pg=PA41&lpg=PA41&dq=Traffic+policing+and+shaping+help+regulate+bandwidth+usage+by+limiting+the+amount+of+traffic&source=bl&ots=YmvtFVj8Op&sig=WrlLCb3w2fEXwSFaBJZnQJ1BCk4&hl=en&sa=X&ved=0ahUKTwig-dOx_tbMAhVHRY8KHV4VB9EQ6AEIKDAC#v=onepage&q=Traffic%20policing%20and%20shaping%20help%20regulate%20bandwidth%20usage%20by%20limiting%20the%20amount%20of%20traffic&f=false)

**NEW QUESTION: 110**

What are the two benefits of using SIP dial rules on an IP phone? (Choose two.)

- A. The phone can initiate dialing without sending any signaling messages to Cisco Unified Communications Manager.
- B. If Cisco Unified Communications Manager requires more digits, another en bloc message is used to send the remaining digits from the SIP.
- C. If Cisco Unified Communications Manager requires more digits, KPML can be used to send the remaining digits from the SIP phone to Cisco Unified Communications Manager one-by-one.
- D. If dialed digits match an entry of a SIP dial rule, the dialed string is sent in a single SIP 200 OK message to Cisco Unified Communications Manager.
- E. The phone can detect invalid numbers and play a reorder tone without sending any signaling messages to Cisco Unified Communications Manager.

Answer: ([SHOW ANSWER](#))

#### NEW QUESTION: 111

Out of the route pattern configuration parameters listed below, which one of these automatically sets the call classification to on-net when you deselect the check box?

- A. Force Off-Net
- B. External Called Number Mask
- C. Pre-Dot Digit Manipulation
- D. Provide Outside Dial Tone

Answer: ([SHOW ANSWER](#))

#### NEW QUESTION: 112

Refer to the exhibit.

| Number Type          | Prefix | Strip Digits | Calling Search Space |
|----------------------|--------|--------------|----------------------|
| National Number      |        | 0            | <None>               |
| International Number | +      | 2            | Trans_CLID_CSS       |
| Unknown Number       |        | 0            | <None>               |
| Subscriber Number    | +1408  | 0            | Trans_CLID_CSS       |

Trans\_CLID\_CSS Match pattern: \\+4922.!  
DDI: Pre Dot - Prefix +4912

Device Pool CSS - Default

What is the output displayed on the IP phone when the incoming calling number is 00492288224002 and has a TON = unknown?

- A. +49122288224002
- B. 00492288224002
- C. +492288224002
- D. 491288224002
- E. +49128822400

Answer: B ([LEAVE A REPLY](#))

#### NEW QUESTION: 113

Which two options list the transcoder types that Cisco Unified Communications Manager supports?

- A. PLDM and SER Termination Point
- B. IOS and media termination point software and hardware
- C. IOS enhanced media termination point and WS-SVC-CMM
- D. virtual media termination point and codec termination point
- E. transcoding and ISR termination point

Answer: ([SHOW ANSWER](#))

Explanation/Reference:

Reference: [https://www.cisco.com/c/en/us/td/docs/voice\\_ip\\_comm/cucm/admin/10\\_0\\_1/ccmsys/CUCM\\_BK\\_SE5FCFB6\\_00\\_cucm-system-guide-100/CUCM\\_BK\\_SE5FCFB6\\_00\\_cucm-system-guide-100\\_chapter\\_011001.pdf](https://www.cisco.com/c/en/us/td/docs/voice_ip_comm/cucm/admin/10_0_1/ccmsys/CUCM_BK_SE5FCFB6_00_cucm-system-guide-100/CUCM_BK_SE5FCFB6_00_cucm-system-guide-100_chapter_011001.pdf)

#### NEW QUESTION: 114

Which statement about translation patterns and transformation patterns is true?

- A. Transformation patterns support numeric and alphanumeric values, whereas translation patterns support only numeric values.
- B. Both transformation patterns and translation patterns support numeric and alphanumeric values.
- C. Both transformation patterns and translation patterns support only numeric values.
- D. Translation patterns support numeric and alphanumeric values, whereas transformation patterns support only numeric values.

Answer: (SHOW ANSWER)

#### NEW QUESTION: 115

Refer to the exhibit.

```
PRI Channel Statistics:
ISDN Se2/0/0:23, Channel [1-24]
Configured Isdn Interface (dsl) 0
Channel State(0=Idle 1=Proposed 2=Busy 3=Reserved 4=Restart
5=Maint_Pend)
Channel : 1 2 3 4 5 6 7 8 9 0 1 2 3 4 5 6 7 8 9 0 1 2 3 4
State : 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 3
Service State (0=Inservice 1=Maint 2=Outofservice 8=MaintPend
9=OOSPend)
Channel : 1 2 3 4 5 6 7 8 9 0 1 2 3 4 5 6 7 8 9 0 1 2 3 4
State : 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 2
```

An engineer is troubleshooting an issue with an ISDN Primary Rate Interface. Which command produces this output?

- A. show isdn service
- B. show isdn history
- C. show isdn active
- D. show isdn status

Answer: A (LEAVE A REPLY)

#### NEW QUESTION: 116

The Cisco IP Phone of your company operator is on extension 7000. However, users normally dial 0 to get the operator. Which action sends calls that were made to 0 to the operator?

- A. Create a translation pattern for 0 and a called party transform mask of 7000.
- B. Create a translation pattern for 7000 and a called party transform mask of 0.

- C. Create the IP phone overlay feature so the operator's line has 0 and 7000 assigned to it.
- D. Create a route pattern of 0 and use a called party transform mask of 7XXX.
- E. Create a translation pattern for 0 and a calling party transform mask of 7000.

**Answer:** ([SHOW ANSWER](#))

#### NEW QUESTION: 117

Which three items have a calling search space? (Choose three.)

- A. transcoder
- B. gateways
- C. phone lines
- D. phones
- E. partitions
- F. AAR groups

**Answer:** ([SHOW ANSWER](#))

Explanation/Reference:

Reference: [http://www.cisco.com/c/en/us/td/docs/voice\\_ip\\_comm/cucm/admin/7\\_1\\_2/ccmcfg/bccm-712-cm/b03parti.html](http://www.cisco.com/c/en/us/td/docs/voice_ip_comm/cucm/admin/7_1_2/ccmcfg/bccm-712-cm/b03parti.html)

#### NEW QUESTION: 118

You are a Network Engineer at ABC Company. You want to ensure that when the contact center agent for the company makes an external call, the extension 897 654 300 is displayed. How do you resolve this issue?

- A. Configure an external phone number mask on the Directory Number configuration page.
- B. Configure a Line Text Label on the Directory Number configuration page.
- C. Configure an external phone number mask on the Device configuration page.
- D. Configure a translation pattern with the pattern 897654300.

**Answer:** ([SHOW ANSWER](#))

#### NEW QUESTION: 119

A voice engineer is troubleshooting an issue that involves incoming calls to POTS lines. Which command verifies the digits that are received by voice-port?

- A. debug call treatment
- B. debug voice protocol
- C. debug vtsp session
- D. debug voip dial-per
- E. debug vtsp dsp

**Answer:** ([SHOW ANSWER](#))

#### NEW QUESTION: 120

Which protocol can Cisco Unified Communications Manager not use to monitor the status of the gateway?

FreeExam.net

- A. MGCP
- B. SCCP
- C. H.323
- D. SIP

Answer: ([SHOW ANSWER](#))

#### NEW QUESTION: 121

Which three functions does the Master Agent perform? (Choose three.)

- A. Controls backup and restoration of the system
- B. Communicates with archive devices
- C. Interfaces with the DRS agent on the Administrator PC
- D. Executes backup and restore tasks
- E. Stores DRS configuration
- F. Acts as a backup repository

Answer: ([SHOW ANSWER](#))

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#### NEW QUESTION: 122

In Cisco VCS, to enable URI dialing via DNS, you must configure at least one DNS zone. Which path do you take to configure the zone?

- A. Status > Zones
- B. Configuration > Protocols > H.323
- C. Configuration > Protocols > SIP
- D. Configuration > Zones > Zones

Answer: ([SHOW ANSWER](#))

#### NEW QUESTION: 123

If an IP phone loses connectivity with its primary server and registers with its backup server, what happens when connectivity to the primary server is restored?

- A. Once connectivity is re-established with the primary server, the IP phone will wait until there have been three successful TCP keepalive exchanges before it will re-register with the primary server.
- B. Once the IP phone registers with the backup server, the administrator will need to reset the IP phone for it to re-register with the primary server.

C. The IP phone will continuously try to re-establish a connection with the primary server; if successful, the IP phone will re-register with the primary.

D. The IP phone will try to re-establish a connection to the primary server every 90 seconds.

**Answer:** ([SHOW ANSWER](#))

#### **NEW QUESTION: 124**

Which two options are forms of redundancy provided by Cisco Unified Communications Manager? (Choose two.)

A. Voice-mail redundancy

B. Media resource redundancy

C. Emergency responder redundancy

D. CTI redundancy

**Answer:** ([SHOW ANSWER](#))

#### **NEW QUESTION: 125**

The Cisco Unified Communications architecture supports which three technologies? (Choose three.)

A. Physical Security

B. Intelligent Building Management

C. Workforce Optimization

D. Smart Grid Management

E. Cisco Unified Video Advantage

F. Contact Center

**Answer:** C,E,F ([LEAVE A REPLY](#))

#### **NEW QUESTION: 126**

Which three options can be used to configure digit manipulation for dial peers? (Choose three.)

A. digit translation rules

B. called party transformations

C. route partitions

D. forward digits

E. route patterns

F. digit stripping and prefixes

G. calling party transformations

**Answer:** ([SHOW ANSWER](#))

Explanation/Reference:

Reference: [http://www.cisco.com/c/en/us/td/docs/ios/12\\_2/voice/configuration/guide/fvvfax\\_c/vvfpeers.html#wp1244453](http://www.cisco.com/c/en/us/td/docs/ios/12_2/voice/configuration/guide/fvvfax_c/vvfpeers.html#wp1244453)

#### **NEW QUESTION: 127**

The Corporation is experiencing poor, choppy audio quality on voice calls placed across their WAN link to and from Madison.

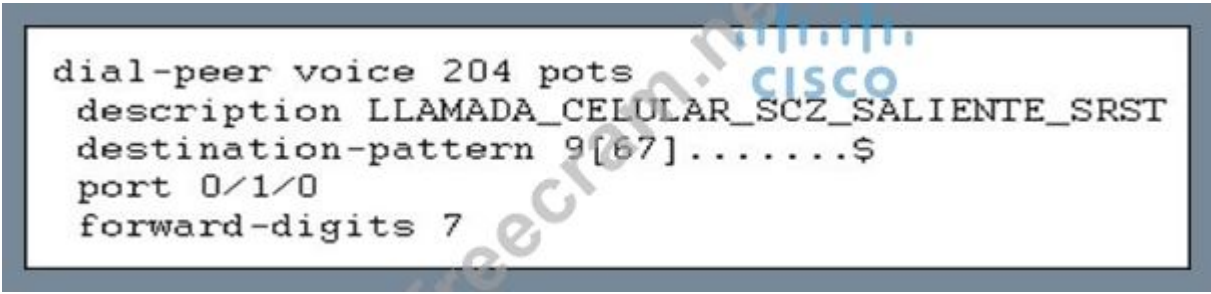
What can be done to the Location parameter for Madison to help alleviate this problem?

- A. Decrease the audio bandwidth setting in the Location configuration window for Madison.
- B. Remove the audio bandwidth parameter in the Location configuration window for Madison.
- C. Increase the audio bandwidth setting in the Location configuration window for Madison.
- D. Nothing, the audio bandwidth Location parameter for Madison is not related to the problem.

Answer: ([SHOW ANSWER](#))

#### NEW QUESTION: 128

Refer to the exhibit.



```
dial-peer voice 204 pots
description LLAMADA_CELULAR_SCZ_SALIENTE_SRST
destination-pattern 9[67].....$
port 0/1/0
forward-digits 7
```

What is the problem when 237 dials 972551915?

- A. The problem is in "forward-digits 7"; the dial peer should be with "forward-digits 8."
- B. The problem is in "forward-digits 7"; the dial peer should be with "forward-digits 5."
- C. The problem is in "forward-digits 7"; the dial peer should be with "forward-digits 2."
- D. The problem is in "forward-digits 7"; the dial peer should be with "forward-digits 9."

Answer: ([SHOW ANSWER](#))

#### NEW QUESTION: 129

Which two options are resources in a route list that route calls that match a defined route pattern?

(Choose two.)

- A. virtual machine servers
- B. route groups
- C. gateways
- D. IP phones

Answer: ([SHOW ANSWER](#))

#### NEW QUESTION: 130

Refer to the exhibit.

**Cisco Unified CM Administration**  
For Cisco Unified Communications Solutions

Navigation: Cisco Unified CM Administration | Search Documents

System ▾ Call Routing ▾ Media Resources ▾ Advanced Features ▾ Device ▾ Application ▾ User Management ▾ Bulk Administration ▾

### Region Configuration

Related Links: Back

Save Delete Reset Apply Config Add New

Name\* REGION1

#### Region Relationships

| Region  | Audio Codec Preference List | Maximum Audio Bit Rate | Maximum Session Bit Rate for Video Calls | Maximum Bit Rate for Video Immersion |
|---------|-----------------------------|------------------------|------------------------------------------|--------------------------------------|
| REGION1 | CCNP COLLAB                 | 60 kbps                | 304 kbps                                 | 2048 kbps                            |
| REGION2 | CCNP COLLAB                 | 64 kbps (G.722, G.711) | Use System Default (364 kbps)            | Use System Default                   |

NOTE: Regions not displayed Use System Default Use System Default Use System Default

**Cisco Unified CM Administration**  
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### Audio Codec Preference List Configuration

Related Links:

Save Delete Copy Add New

#### Status

Status: Ready

#### Audio Codec Preference List Information

Name\* CCNP COLLAB

Description\* CCNP COLLAB

Codecs in List\*  
 G.722 48k  
 G.711 U-Law 64k  
 G.729 8k  
 G.711 A-Law 56k

An engineer is troubleshooting a codec negotiation issue where both endpoints that are involved in the call support the codecs listed in the exhibit. Which audio codec is selected if a call between two endpoints in Region 1 is placed?

- A. G722
- B. G711a
- C. G729
- D. G711u

Answer: ([SHOW ANSWER](#))

#### NEW QUESTION: 131

Refer to the Exhibit. Which phones can Phone 1 reach?



- A. Phone 2
- B. Phone 2, Phone 3, Phone 4 and Phone 5
- C. Phone 5
- D. Phone 2 and Phone 3
- E. Phone 4
- F. Phone 2, Phone 3 and Phone 4
- G. Phone 3
- H. Phone 4 and Phone 5

**Answer:** ([SHOW ANSWER](#))

#### NEW QUESTION: 132

How does Cisco Unified Border Element attach recorder profiles for media forking?

- A. codec configuration
- B. dial peers
- C. SIP trunks
- D. CTI integration

**Answer: B** ([LEAVE A REPLY](#))

Explanation/Reference:

Reference: [https://www.cisco.com/c/en/us/td/docs/ios-xml/ios/voice/cube\\_proto/configuration/xr-3s/cube- proto-xr-3s-book/voi-ntwk-based.html](https://www.cisco.com/c/en/us/td/docs/ios-xml/ios/voice/cube_proto/configuration/xr-3s/cube- proto-xr-3s-book/voi-ntwk-based.html)

#### NEW QUESTION: 133

Which three menu choices will open a window in which you can assign softkey templates? (Choose three.)

- A. Device > Device Settings > Device Profile
- B. System > Enterprise Parameters
- C. System > Device Pool Configuration
- D. System > Device Defaults Configuration
- E. Device > Phone Configuration
- F. Service > Service Parameters

**Answer:** ([SHOW ANSWER](#))

NEW QUESTION: 134

Refer to the exhibit.

CISCO

Cisco Unified CM Administration

For Cisco Unified Communications Solutions

Navigation

System ▾ Call Routing ▾ Media Resources ▾ Voice Mail ▾ Device ▾ Application ▾ User Management ▾ Bulk Administration ▾

SIP Route Pattern Configuration

Related

Save

Delete

Copy

Add New

Status

Add successful

Pattern Definition

Pattern Usage

Enrutamiento de dirección IP

IPv4 Pattern \*

10.200.6.3

IPv6 Pattern

Description

Monterrey

Route Partition

Interina ▾

SIP Trunk/Route List \*

RL-Monterrey ▾

(Edit)

☐ Block Pattern

Calling Party Transformations

☐ Use Calling Party's External Phone Mask

Calling Party Transformation Mask

Prefix Digits (Outgoing Calls)

Calling Line ID Presentation \*

Default ▾

Calling Line Name Presentation \*

Default ▾

Connected Party Transformations

Connected Line ID Presentation \*

Default ▾

Connected Line Name Presentation \*

Default ▾

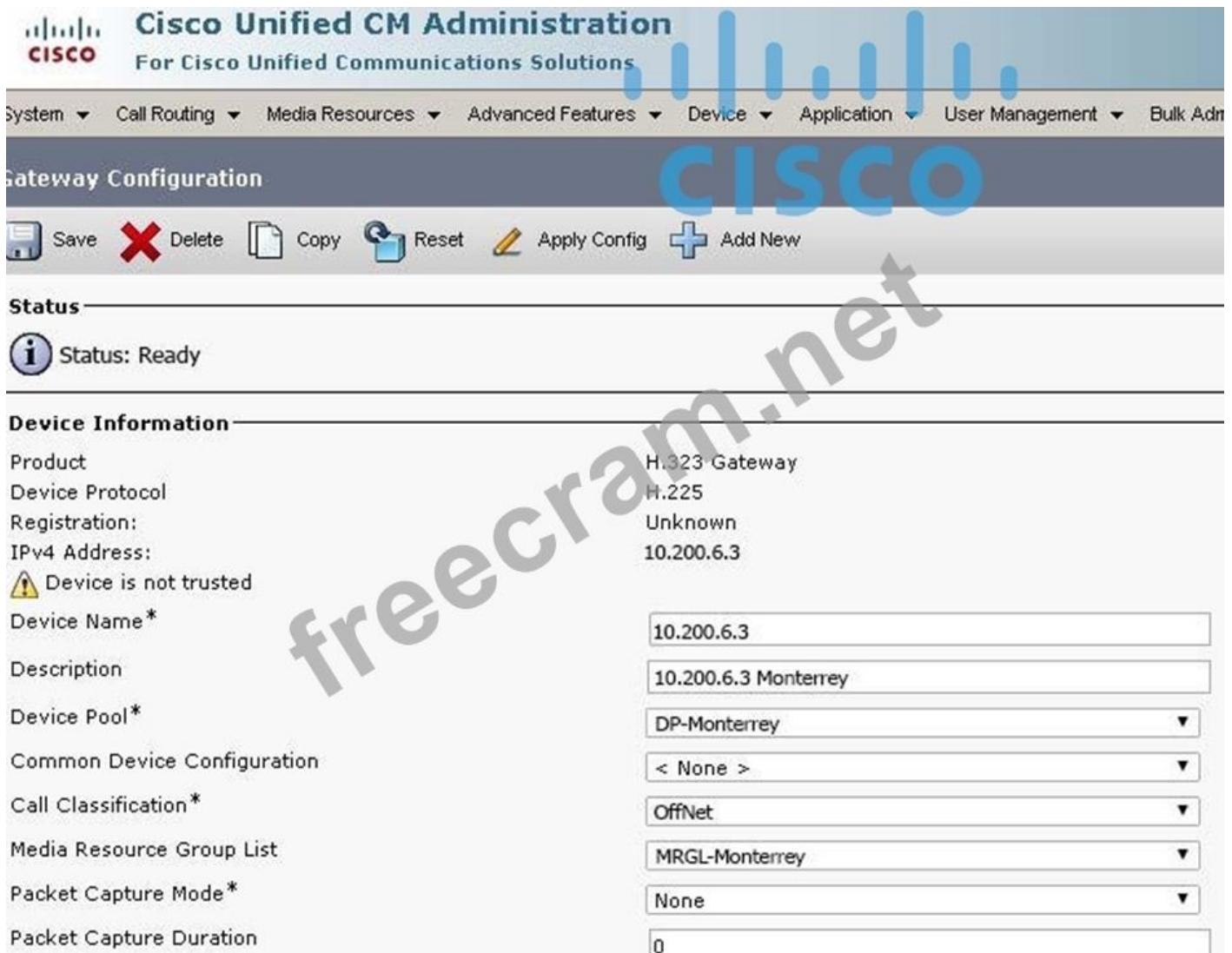
Save

Delete

Copy

Add New

\*- indicates required item.



**Cisco Unified CM Administration**  
For Cisco Unified Communications Solutions

System ▾ Call Routing ▾ Media Resources ▾ Advanced Features ▾ Device ▾ Application ▾ User Management ▾ Bulk Adm

**Gateway Configuration**

Save Delete Copy Reset Apply Config Add New

**Status**  
Status: Ready

**Device Information**

|                             |                      |
|-----------------------------|----------------------|
| Product                     | H.323 Gateway        |
| Device Protocol             | H.225                |
| Registration:               | Unknown              |
| IPv4 Address:               | 10.200.6.3           |
| ⚠ Device is not trusted     |                      |
| Device Name*                | 10.200.6.3           |
| Description                 | 10.200.6.3 Monterrey |
| Device Pool*                | DP-Monterrey ▾       |
| Common Device Configuration | < None > ▾           |
| Call Classification*        | OffNet ▾             |
| Media Resource Group List   | MRGL-Monterrey ▾     |
| Packet Capture Mode*        | None ▾               |
| Packet Capture Duration     | 0                    |

A Network Administrator needs to create a SIP route pattern to communicate with Monterrey's Gateway. The Administrator does not understand why this is not working properly. Which two statements explain the issue and resolve the problem? (Choose two.)

- A.** The Administrator should type the reload command into Monterrey's Gateway
- B.** The issue is caused by a failure of communication between the Monterrey's Gateway and Cisco Unified Communications Manager
- C.** The issue is caused by incorrect protocol integration between Monterrey's Gateway and Cisco Unified Communications Manager.
- D.** The Administrator needs to set up a SIP trunk between Monterrey's Gateway and Cisco Unified Communications Manager.

**Answer:** ([SHOW ANSWER](#))

#### NEW QUESTION: 135

An engineer is designing a new dial plan for a customer that has offices in several countries on four continents around the world. This client also wants to integrate with a Microsoft Lync backend. Which dial plan type should the engineer recommend?

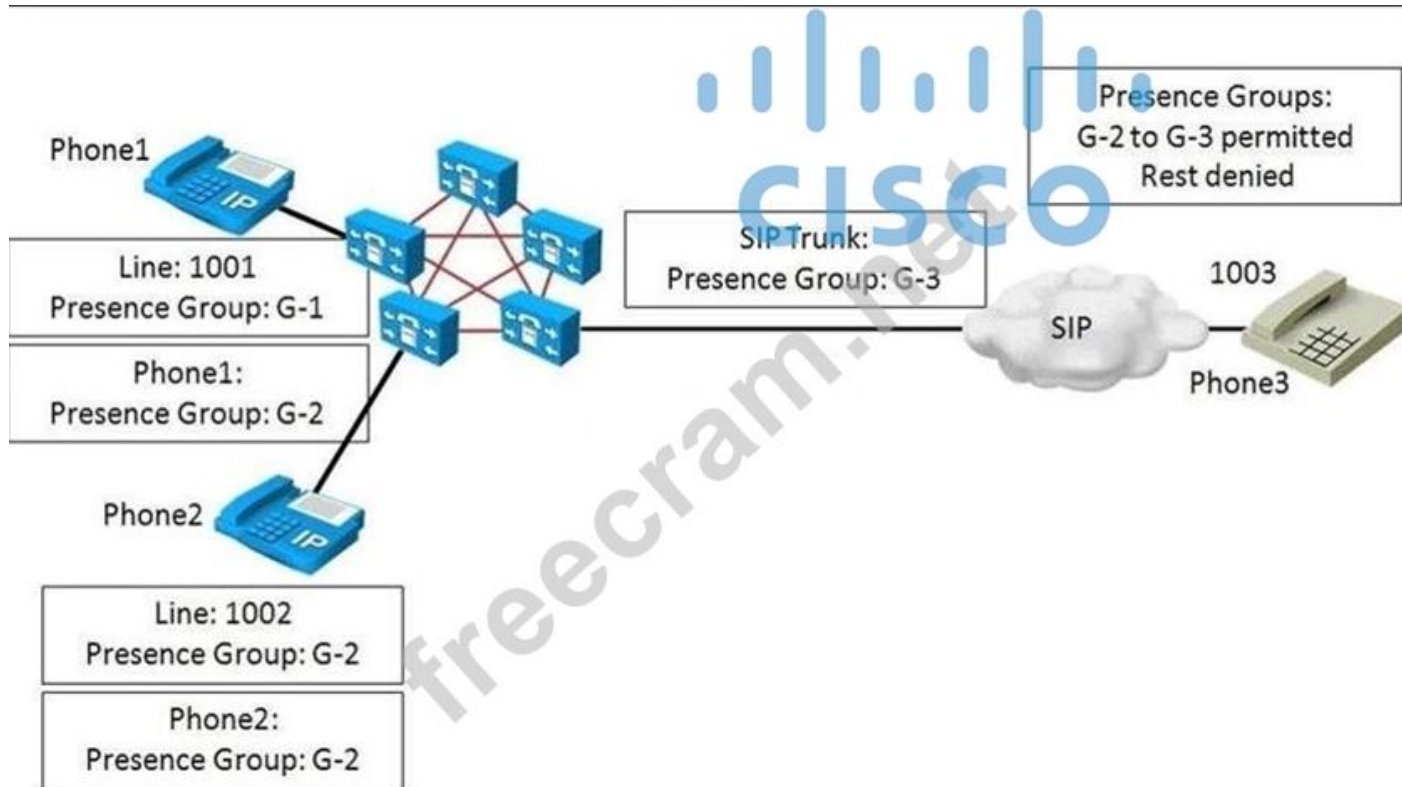
- A.** H.323
- B.** E.164

- C. TEHO
- D. SIP URI

Answer: ([SHOW ANSWER](#))

#### NEW QUESTION: 136

Refer to the Exhibit. Which two of these statements about call lists enabled for Cisco Unified Presence are true? (Choose two.)



- A. 1001 cannot watch 1003
- B. 1001 can watch 1002
- C. 1002 can watch 1001
- D. 1003 can watch 1001
- E. 1002 can watch 1003

Answer: ([SHOW ANSWER](#))

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**NEW QUESTION: 137**

What are two characteristics of a Cisco Unified Communications Manager Express dial plan? (Choose two.)

- A. partitions and calling search spaces
- B. COR
- C. route patterns
- D. dial peers
- E. voice translation rules

**Answer:** ([SHOW ANSWER](#))

**NEW QUESTION: 138**

What is the default option for framing when you configure a T1/E1 controller?

- A. Super Frame
- B. alternate mark inversion
- C. Extended Super Frame
- D. cyclic redundancy check 4
- E. high-density bipolar 3
- F. No-CRC4
- G. binary 8-zero substitution
- H. Australia

**Answer:** ([SHOW ANSWER](#))

Explanation/Reference:

Explanation:

The default setting for T1 framing is esf. The default setting for E1 framing is crc4.

Reference: <http://www.cisco.com/en/US/docs/routers/access/1700/1721/software/feature/guide/t1e11721.html>

**NEW QUESTION: 139**

What is the correct path selection for a call that is routed through a Cisco Unified Communications Manager system?

- A. Route Pattern -> Route List -> Route Group -> Devices
- B. Route Group -> Devices -> Route Pattern -> Route List
- C. Route List -> Route Pattern -> Route Group -> Devices
- D. Devices -> Route Group -> Route List -> Route Pattern

**Answer:** ([SHOW ANSWER](#))

**NEW QUESTION: 140**

If the route patterns 5XXX, 51XX, 513X, and 5134 exist within a call route database, which pattern would be matched if the dialed number was 5124?

- A. 5134
- B. 5XXX

C. 51XX

D. 513X

Answer: ([SHOW ANSWER](#))

#### NEW QUESTION: 141

Which option is used to associate devices with a Cisco Unified Communications Manager group?

A. Device pools

B. Locations

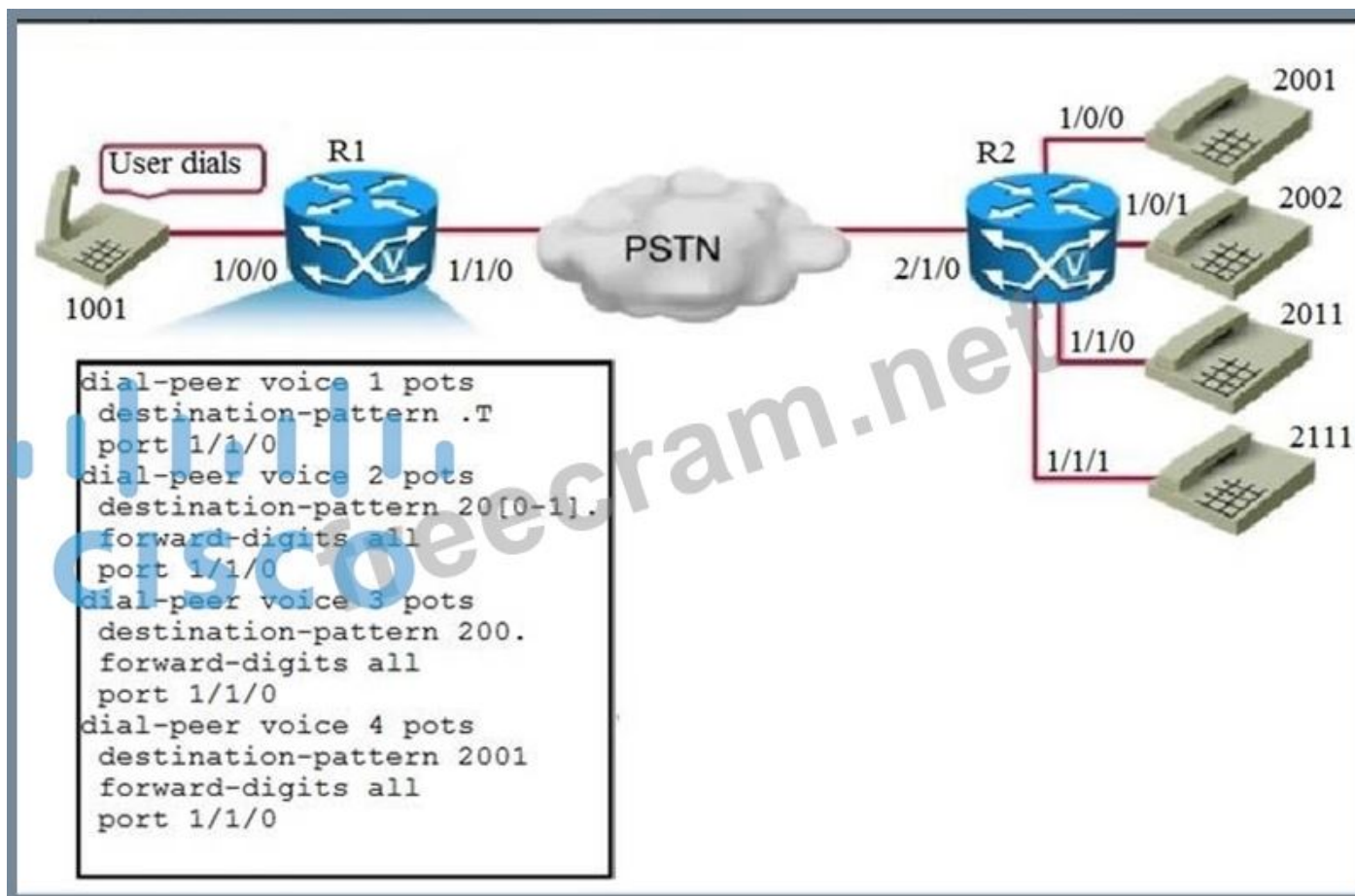
C. Licenses

D. Regions

Answer: ([SHOW ANSWER](#))

#### NEW QUESTION: 142

Refer to Exhibit:



If the user at extension 1001 dials the user at extension 2011, which dial peer is used to route the call?

A. 1

B. 2

C. 4

D. 3

Answer: ([SHOW ANSWER](#))

#### NEW QUESTION: 143

Which two algorithms are valid to distribute calls in a route group? (Choose two.)

- A. circular
- B. bottom-up
- C. round-robin
- D. broadcast
- E. top-down

**Answer:** ([SHOW ANSWER](#))

#### NEW QUESTION: 144

What happens when you try to configure the fourth member of a Cisco Unified CallManager group?

- A. The new member will replace the first member on the list.
- B. Cisco Unified CallManager Administration will display an error message when you attempt to add the fourth member.
- C. Cisco Unified CallManager Administration will display an error and replace the last entered member of the Cisco Unified CallManager group with the new entry.
- D. The fourth member will be added to the sequential list.

**Answer:** ([SHOW ANSWER](#))

#### NEW QUESTION: 145

Refer to the exhibit.

The screenshot shows the 'Disk Format' step of the Cisco TelePresence Conductor installation wizard. On the left, a sidebar lists navigation options: Source, OVF Template Details, End User License Agreement, Name and Location, Deployment Configuration, and Disk Format (which is currently selected and highlighted). Below the sidebar, it says 'Ready to Complete'. The main area displays the 'Datastore' as 'datastore 1' and 'Available space (GB)' as '325.8'. There are three radio button options for disk formatting: 'Thick Provision Lazy Zeroed', 'Thick Provision Eager Zeroed', and 'Thin Provision'. The 'Thick Provision Eager Zeroed' option is selected. A large 'freegram.net' watermark is visible across the center, and the Cisco logo is at the bottom right.

Which option is the most highly recommended option to install Cisco TelePresence Conductor on the virtual machine?

- A. Thick Provision Eager Zeroed
- B. Thick Provision Lazy Zeroed
- C. Thick Provision
- D. Thin Provision

**Answer:** B ([LEAVE A REPLY](#))

Explanation/Reference:

Reference:

[http://www.cisco.com/c/dam/en/us/td/docs/telepresence/infrastructure/conductor/install\\_guide/TelePresence-Conductor-Virtual-Machine-Install-Guide-XC3-0.pdf](http://www.cisco.com/c/dam/en/us/td/docs/telepresence/infrastructure/conductor/install_guide/TelePresence-Conductor-Virtual-Machine-Install-Guide-XC3-0.pdf) (page 8)

**NEW QUESTION: 146**

Which option describes how you add software conference bridges to Cisco Unified Communications Manager?

- A. By installing DSP to a Cisco Unified CM server
- B. By adding a Cisco Unified CM server to the cluster
- C. By adding a software conference bridge using Conference Bridge Configuration
- D. By reassigning other media resources to conference resources

**Answer: B** ([LEAVE A REPLY](#))

**NEW QUESTION: 147**

A customer is integrating two video solutions between regions using a Cisco Unified Border Element and finds that some of the RTP payloads are not compatible with the Cisco Unified Border Element configuration. Which two actions does Cisco recommend to solve this issue? (Choose two.)

- A. Configure a transparent codec on the dial peers.
- B. Configure RTP payload mapping on the dial peers.
- C. Set up a voice class for the video codec.
- D. Configure asymmetric payload on the voice service.
- E. Specify the video payload on the voice service.

**Answer: (**[SHOW ANSWER](#)**)**

**NEW QUESTION: 148**

When the Cisco IOS router or gateway receives a call setup request, what is the PID value of the default dial peer?

- A. PID:3
- B. PID:6
- C. PID:0
- D. PID:4
- E. PID:5
- F. PID:1
- G. PID:2
- H. PID:7

**Answer: (**[SHOW ANSWER](#)**)**

Explanation/Reference:

Reference: <http://www.cisco.com/c/en/us/support/docs/voice/call-routing-dial-plans/14074-in-dial-peer-match.html>

**NEW QUESTION: 149**

Hackers are sending calls through a gateway router that is running SIP. The engineer blocks all SIP messages to the router except from their SIP server, which has an IP address of 208.177.10.1. The customer also must connect via SIP to their Cisco Expressway server with an IP address of 10.1.10.110.

Which series of commands secures the router from the hackers?

**A.** voice service voip

voip security allow

ipv4 208.177.10.1 255.255.255.255

ipv4 10.1.10.110 255.255.255.255

**B.** voice service voip

ip address trusted list

ipv4 208.177.10.1 255.255.255.255

ipv4 10.1.10.110 255.255.255.255

**C.** voice service voip

allow-connections sip to sip

ipv4 208.177.10.1 255.255.255.255

ipv4 10.1.10.110 255.255.255.255

**D.** voice service voip

address-hiding

ipv4 208.177.10.1 255.255.255.255

ipv4 10.1.10.110 255.255.255.255

**Answer:** ([SHOW ANSWER](#))

Explanation/Reference:

Reference: <http://www.cisco.com/c/en/us/support/docs/voice/call-routing-dial-plans/112083-tollfraud-ios.html>

### NEW QUESTION: 150

Which two options can be used to configure digit manipulation for dial peers? (Choose two.)

**A.** called party transformations

**B.** digit translation rules

**C.** route partitions

**D.** forward digits

**E.** route patterns

**Answer:** ([SHOW ANSWER](#))

### NEW QUESTION: 151

What is the correct configuration order of call routing to support off-net call routing?

**A.** device, route group, route list, route pattern

**B.** route list, route group, route pattern, device

**C.** route pattern, route list, route group, device

**D.** device, route list, route group, route pattern

**Answer:** ([SHOW ANSWER](#))

Explanation/Reference:

Reference: <https://books.google.com.pk/books?id=0DOXAwAAQBAJ&pg=PA109&lpg=PA109&dq=off-net>

+call+routing+route+pattern,+route+list,+route+group,  
+device&source=bl&ots=YmvtFVjaRi&sig=c3gBcT3X73RI9J1Sbm5n7d1bLeI&hl=en&sa=X&ved=0ahUKE  
wiFpbac\_9bMAhWKp48KHdBMA0sQ6AEIPzAF#v=onepage&q=off-net%20call%20routing%20route%  
20pattern%2C%20route%20list%2C%20route%20group%2C%20device&f=false

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#### NEW QUESTION: 152

Which codec provides native WebRTC-based video chat support in browsers such as Mozilla Firefox?

- A. H.265
- B. WMV
- C. H.264
- D. G.711
- E. H.323

Answer: ([SHOW ANSWER](#))

#### NEW QUESTION: 153

Which two types of encoding are available to configure a T1 controller for CAS? (Choose two.)

- A. B8SF
- B. AMI
- C. B8ZS
- D. SF
- E. AME
- F. ESF

Answer: ([SHOW ANSWER](#))

Explanation/Reference:

Reference: <http://www.cisco.com/c/en/us/support/docs/voice/digital-cas/24642-cas-config.html>

#### NEW QUESTION: 154

What are the characteristics of a Cisco Unified Communications Manager group?

- A. A Cisco Unified Communications Manager group is a prioritized list of (one or more) call-processing and application servers.
- B. Each device has to have a Cisco Unified Communications Manager group assigned, which will determine the primary and backup servers to which it can register.

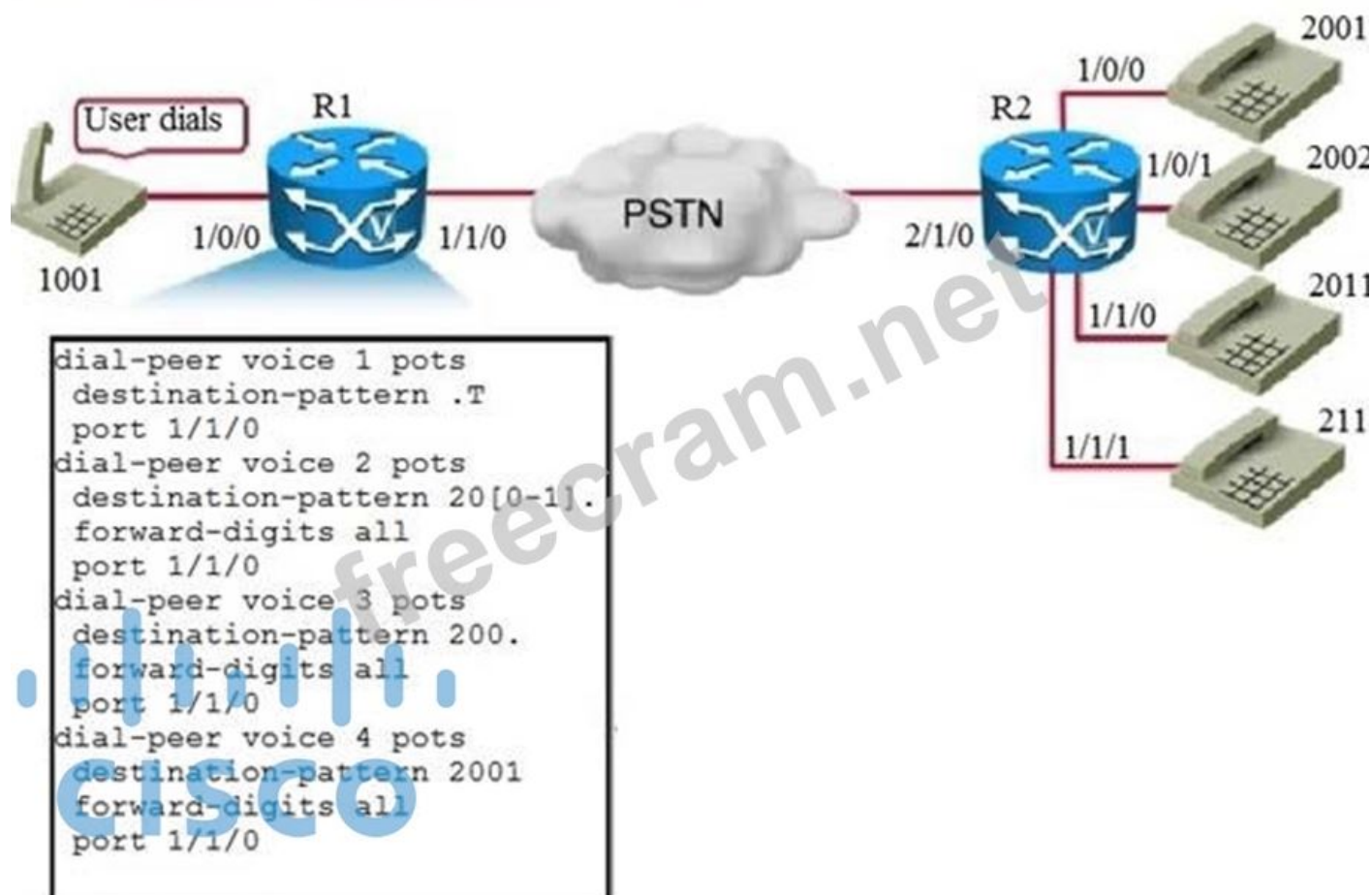
C. Each call-processing server can be assigned to a single Cisco Unified Communications Manager group.

D. A single Cisco Unified Communications Manager group can exist in a single cluster.

Answer: ([SHOW ANSWER](#))

#### NEW QUESTION: 155

Refer to Exhibit:



If the user at extension 1001 dials the user at extension 2002, which dial peer is used to route the call?

A. 3

B. 4

C. 2

D. 1

Answer: ([SHOW ANSWER](#))

#### NEW QUESTION: 156

Which device uses an MTP to insert itself into and force media through a call flow?

A. conference resource

B. TRP

C. SCCP

D. H 323 gateway

Answer: ([SHOW ANSWER](#))

Explanation/Reference:

Reference: [https://www.cisco.com/c/en/us/td/docs/voice\\_ip\\_comm/cucm/srnd/collab10/collab10/media.html](https://www.cisco.com/c/en/us/td/docs/voice_ip_comm/cucm/srnd/collab10/collab10/media.html)

#### NEW QUESTION: 157

Which two tasks should be performed before you delete a phone button template that is currently in use? (Choose two.)

- A. Assign a different phone button template to any devices that use the phone button template that you want to delete.
- B. Check the "ALLOW WITH CONFIRMATION" checkbox in the setting "Delete Button Template in use" under System Parameters, and save the setting.
- C. Rename the button template as "OUT OF SERVICE" to dissociate it from any devices that use the phone button template that you want to delete.
- D. Delete the devices that use the phone button template that you want to delete.

Answer: ([SHOW ANSWER](#))

#### NEW QUESTION: 158

Refer to the exhibit.

| Call type description |        |               | Screen licenses required per call | Maximum calls by core size (with licenses providing 100% of capacity) |                         |
|-----------------------|--------|---------------|-----------------------------------|-----------------------------------------------------------------------|-------------------------|
| Main video            | Audio  | Content       |                                   | 8-core Virtual Machine                                                | 10-core Virtual Machine |
|                       |        |               |                                   | 4 screen licenses                                                     | 6 screen licenses       |
| -                     | Mono   | -             | 1/52                              | 104*                                                                  | 104*                    |
| 360p30+               | Mono   | In main video | 1/8                               | 33                                                                    | 49                      |
| 480p30                | Mono   | In main video | 1/4                               | 16                                                                    | 24                      |
| 480p30                | Stereo | 720p5         | 1/3                               | 12                                                                    | 18                      |

How many licenses do you need to operate three calls to 480p30 Audio Stereo, using Cisco TelePresence Server on Virtual Machine 3.1?

- A. 1/4 license
- B. 1 license

C. 1/52 license

D. 1/3 license

Answer: (SHOW ANSWER)

### NEW QUESTION: 159

What is the recommended QoS marking for desktop video endpoints?

A. AF31

B. DSCP 46

C. AF41

D. DSCP 26

Answer: (SHOW ANSWER)

### NEW QUESTION: 160

Refer to the exhibit.

The top screenshot shows the 'Region Configuration' page in Cisco Unified CM Administration. It includes a navigation bar with tabs like System, Call Routing, Media Resources, Advanced Features, Device, Application, User Management, and Bulk Administration. Below the navigation bar, there are icons for Save, Delete, Reset, Apply Config, and Add New. The 'Region Information' section shows a text field for 'Name\*' with the value 'REGION1'. The 'Region Relationships' section contains a table with the following data:

| Region  | Audio Codec Preference List | Maximum Audio Bit Rate | Maximum Session Bit Rate for Video Calls | Maximum |
|---------|-----------------------------|------------------------|------------------------------------------|---------|
| REGION1 | CCNP COLLAB                 | 60 kbps                | 384 kbps                                 |         |
| REGION2 | CCNP COLLAB                 | 64 kbps (G.722, G.711) | Use System Default (384 kbps)            |         |

Below the table, there is a note: 'NOTE: Regions not displayed' and four 'Use System Default' links.

The bottom screenshot shows the 'Audio Codec Preference List Configuration' page. It includes the same navigation bar and icons. The 'Status' section shows 'Status: Ready'. The 'Audio Codec Preference List Information' section shows a text field for 'Name\*' with the value 'CCNP COLLAB', a text field for 'Description\*' with the value 'CCNP COLLAB', and a list box for 'Codecs in List\*' containing the following items: G.722 48k, G.711 U-Law 64 k, G.729 8k, and G.711 A-Law 56k.

An engineer is troubleshooting this video conference issue:

A video call between a Cisco 9971 in Region1 and another Cisco 9971 in Region1 works.

As soon as the Cisco 9971 in Region1 conferences in a Cisco 8945 in Region2, the Region1 endpoint cannot see the Region2 endpoint video.

Which cause of this issue is true?

- A. Maximum Audio Bit Rate should be increased.
- B. Cisco 8945 does not have a camera connected.
- C. Maximum Session Bit Rate for Video Calls is too low.
- D. Maximum Session Bit Rate for Immersive Video Calls is too low.

**Answer:** ([SHOW ANSWER](#))

#### **NEW QUESTION: 161**

Every VoIP endpoint must select one specific codec. Which of these codecs is the mandatory codec?

- A. G.722
- B. H.264
- C. G.711
- D. G.728
- E. G.729

**Answer:** ([SHOW ANSWER](#))

#### **NEW QUESTION: 162**

Which two options are IOS conference bridge types in CUCM? (Choose two.)

- A. Cisco IOS Conference Bridge
- B. CiscoIOS Hardware Conference Bridge
- C. CiscoIOS Software Conference Bridge
- D. CiscoIOS Standard Conference Bridge
- E. CiscoIOS Enhanced Conference Bridge

**Answer:** ([SHOW ANSWER](#))

#### **NEW QUESTION: 163**

An engineer is configuring a connection between Cisco Unified Communications Manager and Cisco VCS Control for a company that has video call control handled by Cisco UCM. Which zone must be created in VCS Control?

- A. traversal server
- B. DNS
- C. traversal client
- D. neighbor

**Answer:** ([SHOW ANSWER](#))

#### **NEW QUESTION: 164**

Which discard digits instruction removes the access code from a number before passing the number onto an adjacent system?

- A. PreAt
- B. Trailing-#
- C. NoDigits
- D. PreDot

Answer: ([SHOW ANSWER](#))

#### NEW QUESTION: 165

Which two Cisco Unified CallManager tasks are required to route calls from Cisco Unified CallManager to the PSTN via an H.323 gateway? (Choose two.)

- A. Add PSTN route patterns pointing to the gateway
- B. Add a gatekeeper from the Device menu
- C. Configure the voice endpoints in Cisco Unified CallManager
- D. Configure the IP phones with an external phone number mask
- E. Add an H.323 gateway from the Device menu
- F. Configure the signaling protocol used on the voice circuit

Answer: ([SHOW ANSWER](#))

#### NEW QUESTION: 166

When comparing SIP H.323 and MGCP, which of the following is specific to SIP?

- A. SIP defines Translation per the gateway.
- B. SIP provides support for third-party entities.
- C. SIP has a centralized dial plan.
- D. The dial plan is directly on the gateway.

Answer: ([SHOW ANSWER](#))

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#### NEW QUESTION: 167

What are two characteristics of a Cisco Unified Communications Manager Enterprise parameter? (Choose two.)

- A. It is used to configure settings for specific services.
- B. It sets initial values of device-defaults.

- C. It is used to define end-user authentication settings.
- D. It is used to define clusterwide settings.
- E. It is used to configure settings for different services.

**Answer: B,D ([LEAVE A REPLY](#))**

#### **NEW QUESTION: 168**

Which statement about creating Cisco IP Phone Services is true?

- A. Cisco IP Phone Services must have a unique name.
- B. Cisco IP Phone Services must have the name "Service" in the description.
- C. You must use hostnames in the Cisco IP Phone Service URL.
- D. You can create multiple Cisco IP Phone Services with the same name.
- E. You must use IP addresses in the Cisco IP Phone Service URL.

**Answer: ([SHOW ANSWER](#))**

#### **NEW QUESTION: 169**

What are two characteristics that are associated with the End User and User account types? (Choose two.)

- A. They cannot use LDAP.
- B. They are used to authorize applications.
- C. They can utilize LDAP.
- D. The Application is associated with the End User and User account types.
- E. They can be listed in a directory.

**Answer: ([SHOW ANSWER](#))**

#### **NEW QUESTION: 170**

Which two of these describe Meet-Me conferencing? (Choose two.)

- A. Anyone who has calling privileges to call the directory number while the conference is active can join the conference.
- B. A range of directory numbers must be allocated for exclusive use by the conference.
- C. Multiple ad hoc conference can be added.
- D. The conference originator of the conference acts as the controller of the conference and is the only participant who can add or remove other participants.
- E. Any participant can add and remove other conference participants.

**Answer: ([SHOW ANSWER](#))**

#### **NEW QUESTION: 171**

In a centralized call-processing solution, there are five sites connected through an IP WAN. To optimize the utilization of the IP WAN, CAC needs to be implemented. How should CAC be implemented?

- A. Use locations CAC with a single location
- B. Use a gatekeeper to only control those locations that use H.323 gateways and a Cisco Unified Border Element to control those sites that use MGCP or SIP gateways

- C. Use locations CAC with five locations in addition to Hub none
- D. Use a gatekeeper to control allocated bandwidth properly
- E. Use a gatekeeper and Cisco Unified Border Element to provide CAC to sites that use a combination of SIP and MGCP gateways

**Answer:** ([SHOW ANSWER](#))

#### **NEW QUESTION: 172**

Which two options can a route pattern be assigned? (Choose two.)

- A. a gateway that is not assigned to a route group
- B. a gateway that is assigned to a route group
- C. a route list with no route groups configured
- D. a route list with one or more route groups configured

**Answer:** ([SHOW ANSWER](#))

Explanation/Reference:

Reference: [http://www.cisco.com/c/en/us/td/docs/voice\\_ip\\_comm/cucm/admin/7\\_1\\_2/ccmsys/accm-712-cm/a03rp.html](http://www.cisco.com/c/en/us/td/docs/voice_ip_comm/cucm/admin/7_1_2/ccmsys/accm-712-cm/a03rp.html) (Route Plan Overview)

#### **NEW QUESTION: 173**

Which result of this translation rule is true?

Voice translation-rule 1

rule 1 /.\*/ /1234/

- A. Any dialed digits are translated to 1234.
- B. 1234 is translated to 123.
- C. 1234 is translated to 234.
- D. 1234 is prepended to original dialed number.

**Answer:** ([SHOW ANSWER](#))

#### **NEW QUESTION: 174**

Which of these media resources can be configured in Cisco Unified Communication manager? (Choose three)

- A. Auto Attendant
- B. Conference Bridge
- C. Transcoder
- D. Voice Termination Point DSP
- E. MOH Server

**Answer:** ([SHOW ANSWER](#))

#### **NEW QUESTION: 175**

When Cisco Unified Communications Manager software is installed on a Cisco MCS server, which level of access does it provide for the customer to the operating system of the server?

- A. Access only through an interface controlled by Cisco TAC

- B. No access.
- C. CLI and GUI access.
- D. GUI access only
- E. CLI access only

**Answer:** ([SHOW ANSWER](#))

#### **NEW QUESTION: 176**

A customer recently installed a Cisco TelePresence MCU 5320. The customer wants a virtual meeting room that restricts participants from communicating before the customer joins the call. Which feature in the MCU enables this functionality?

- A. PIN codes
- B. numeric ID
- C. owner
- D. floor and chair control

**Answer:** ([SHOW ANSWER](#))

#### **NEW QUESTION: 177**

Which two characteristics of a video stream require the implementation of QoS? (Choose two.)

- A. small packets
- B. constant bandwidth
- C. high packet rate
- D. codec used
- E. variable packet rate

**Answer:** ([SHOW ANSWER](#))

#### **NEW QUESTION: 178**

Which two components must an engineer configure before configuring SIP route pattern? (Choose two.)

- A. route filter
- B. SIP trunk
- C. route partition
- D. calling search space
- E. SIP profile

**Answer:** ([SHOW ANSWER](#))

Explanation/Reference:

Reference: [http://www.cisco.com/c/en/us/td/docs/voice\\_ip\\_comm/cucm/admin/8\\_6\\_1/ccmcfg/bccm-861-cm/b03siprp.html](http://www.cisco.com/c/en/us/td/docs/voice_ip_comm/cucm/admin/8_6_1/ccmcfg/bccm-861-cm/b03siprp.html)

#### **NEW QUESTION: 179**

The Ajax Corporation is designing an IP telephony network using Cisco MCS 7845 Series servers, each one capable of supporting 7500 devices. The design must meet these requirements:  
be cost-effective

support up to 7500 phones

provide a minimal level of redundancy

Which configuration will meet Ajax Corporation needs?

**A.** five Cisco Unified Communications Manager servers 1 publisher

1 TFTP server

1 primary subscriber

2 backup subscribers

**B.** four Cisco Unified Communications Manager servers:

1 publisher

1 TFTP server

1 primary subscriber

1 backup subscriber

**C.** three Cisco Unified Communications Manager servers:

1 publisher and TFTP server combined

1 primary subscriber

1 backup subscriber

**D.** two Cisco Unified Communications Manager servers:

1 publisher and TFTP server combined

1 primary subscriber

**Answer:** ([SHOW ANSWER](#))

#### NEW QUESTION: 180

A phone is assigned to a partition that is included in the CSS HQ-Phone. This partition contains a route pattern to block calls to service numbers 1-400-xxx-xxxx. When the engineer makes a test call to the number 1-400-400-9000 from that phone, the call goes through. What is the reason for this issue?

**A.** The route pattern for the number to be blocked is in "NULL" partition.

**B.** The route pattern is misconfigured.

**C.** The route pattern for the number to be blocked is in partition not accessible to the CSS.

**D.** The correct route list is not selected in the route pattern for the dialed number to be blocked.

**E.** The correct gateway is not selected for the route list for the route pattern for the dialed number to be blocked.

**Answer:** ([SHOW ANSWER](#))

#### NEW QUESTION: 181

Where can an engineer configure an IP address for Cisco VCS rendezvous conference on TS Conductor?

**A.** Configuration > Locations

**B.** System > Network interfaces

**C.** Configuration > SIP Settings

**D.** System > DNS

**Answer:** ([SHOW ANSWER](#))

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