

350-801^{Q&As}

Implementing and Operating Cisco Collaboration Core Technologies
(CLCOR)

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QUESTION 1

Which call flow matches traffic from a Mobile and Remote Access registered endpoint to central call control?

- A. Endpoint > Expressway-E > Expressway-C > Cisco Unified CM
- B. Endpoint > Expressway-E > Cisco Unified CM
- C. Endpoint > Expressway-C > Cisco Unified CM
- D. Endpoint > Expressway-C > Expressway-E > Cisco Unified CM

Correct Answer: A

QUESTION 2

What is the major difference between the two possible Cisco IM and Presence high-availability modes?

- A. Balanced mode provides user load balancing and user failover in the event of an outage. Active/standby mode provides an always on standby node in the event of an outage, and it also provides load balancing.
- B. Balanced mode provides user load balancing and user failover only for manually generated failovers. Active/standby mode provides an unconfigured standby node in the event of an outage, but it does not provide load balancing.
- C. Balanced mode provides user load balancing and user failover in the event of an outage. Active/standby mode provides an always on standby node in the event of an outage, but it does not provide load balancing.
- D. Balanced mode does not provide user load balancing, but it provides in the event of an outage. Active/standby mode provides an always on standby node in the event of an outage, but it does not provide load balancing.

Correct Answer: C

Balanced mode provides user load balancing and user failover in the event of an outage. Active/standby mode provides an always on standby node in the event of an outage, but it does not provide load balancing.

Here is a more detailed explanation of the two modes:

Balanced mode: In balanced mode, the IM and Presence Service nodes are configured to work together to provide high availability. The nodes are configured in a redundancy group, and the system automatically balances the load of users

across the nodes in the group. If one of the nodes fails, the system automatically fails over the users to the other nodes in the group. **Active/standby mode:** In active/standby mode, one of the IM and Presence Service nodes is designated as

the active node, and the other nodes are designated as standby nodes. The active node handles all of the user traffic, and the standby nodes are only used if the active node fails. If the active node fails, the system automatically fails over to

one of the standby nodes.

QUESTION 3

Which two DNS records must be created to configure Service Discovery for on-premises Jabber? (Choose two.)

- A. `_cisco-uds._tls.cisco.com` pointing to the IP address of Cisco Unified Communications Manager
- B. `_cuplogin._tcp.cisco.com` pointing to a record of IMandP
- C. `_cuplogin._tls.cisco.com` pointing to the IP address of IMandP
- D. `_cisco-uds._tcp.cisco.com` pointing to a record of Cisco Unified CM
- E. `_xmpp._tls.cisco.com` pointing to a record of IMandP

Correct Answer: BD

`_cisco-uds._tcp.cisco.com` pointing to a record of Cisco Unified CM `_cuplogin._tcp.cisco.com` pointing to a record of IMandP

Private DNS SRV Records for CUCM and CUCM IMP Clusters ServiceProtocolDomainsPriorityWeightPortTarget_cisco-uds._tcp.Cisco.com10108443cucm1.cisco.com_ciscouds._tcp.Cisco.com10108443cucm2.cisco.com_cuplogin._tcp.Cisco.com10105061imp1.cisco.com_cuplogin._tcp.Cisco.com10105061imp2.cisco.com

Reference: https://www.cisco.com/c/en/us/td/docs/voice_ip_comm/jabber/Windows/9_7/CJAB_BK_C606D8A9_00_cisco-jabber-dns-configuration-guide/CJAB_BK_C606D8A9_00_cisco-jabber-dns-configuration-guide_chapter_010.html

QUESTION 4

Which version is used to provide encryption for SNMP management traffic in collaboration deployments?

- A. SNMPv1
- B. SNMPv3
- C. SNMPv2
- D. SNMPv2c

Correct Answer: B

Reference: <https://www.cisco.com/c/en/us/td/docs/ios-xml/ios/snmp/configuration/xs-16/snmp-xe-16-book/nm-snmp-encrypt-snmp-support.html>

QUESTION 5

What is a capability of the call forwarding feature in a Cisco Webex dial plan?

- A. device pool selection
- B. Call Admission Control
- C. business continuity

D. ringtone selection

Correct Answer: C

Call forwarding is a feature that allows users to forward incoming calls to another number. This can be useful in a number of situations, such as when a user is not available to take a call, or when a user wants to forward calls to a different

number during certain times of the day.

Call forwarding can be used to improve business continuity by ensuring that calls are always answered, even if the user is not available. For example, if a user is out of the office, they can forward their calls to their voicemail or to another

employee. This ensures that customers and clients can always reach someone, even if the user is not available.

QUESTION 6

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

- A. Configure a translation pattern that has a Calling Party Transform Mask of XXXXXXXXXX.
- B. On the inbound SIP trunk, change Significant Digits to 10.
- C. Change the service parameter Apply Transformations On Remote Number to True.
- D. Configure a calling party transformation pattern that keeps only the last 10 digits.

Correct Answer: D

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

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

QUESTION 7

What are two QoS requirements for VoIP traffic? (Choose two.)

- A. Loss must be no more than 1 percent.
- B. Voice traffic must be marked to DSCP AF41.
- C. One-way latency must be no more than 200 ms.
- D. Voice traffic must be marked to DSCP EF.
- E. Average one-way jitter is greater than 50 ms.

Correct Answer: AD

QUESTION 8

Which datastore and protocol is used for saving back-up files within the Disaster Recovery System of Cisco UCM?

- A. remote disk on a CIFS share
- B. remote disk on the SFTP server
- C. local disk on the Cisco UCM server
- D. remote disk on an NFS share

Correct Answer: B

-A user interface for performing backup and restore tasks;

-A distributed system architecture for performing backup and restore functions, including monitoring the current backup status and providing a history log;

-Scheduled or manual backups;

-Backups archived to a physical drive or remote SFTP server;

QUESTION 9

Users want their mobile phones to be able to access their cisco unity connection mailboxes with only having to enter their voicemail pin at the login prompt calling pilot number where should an engineer configure this feature?

- A. message settings
- B. greetings
- C. alternate extensions
- D. transfer rules

Correct Answer: C

QUESTION 10

As a voice engineer, which two recommendations do you to make to your company to optimize Cisco Unified Communications Manager configuration to reduce the number of toll fraud incidents? (Choose two.)

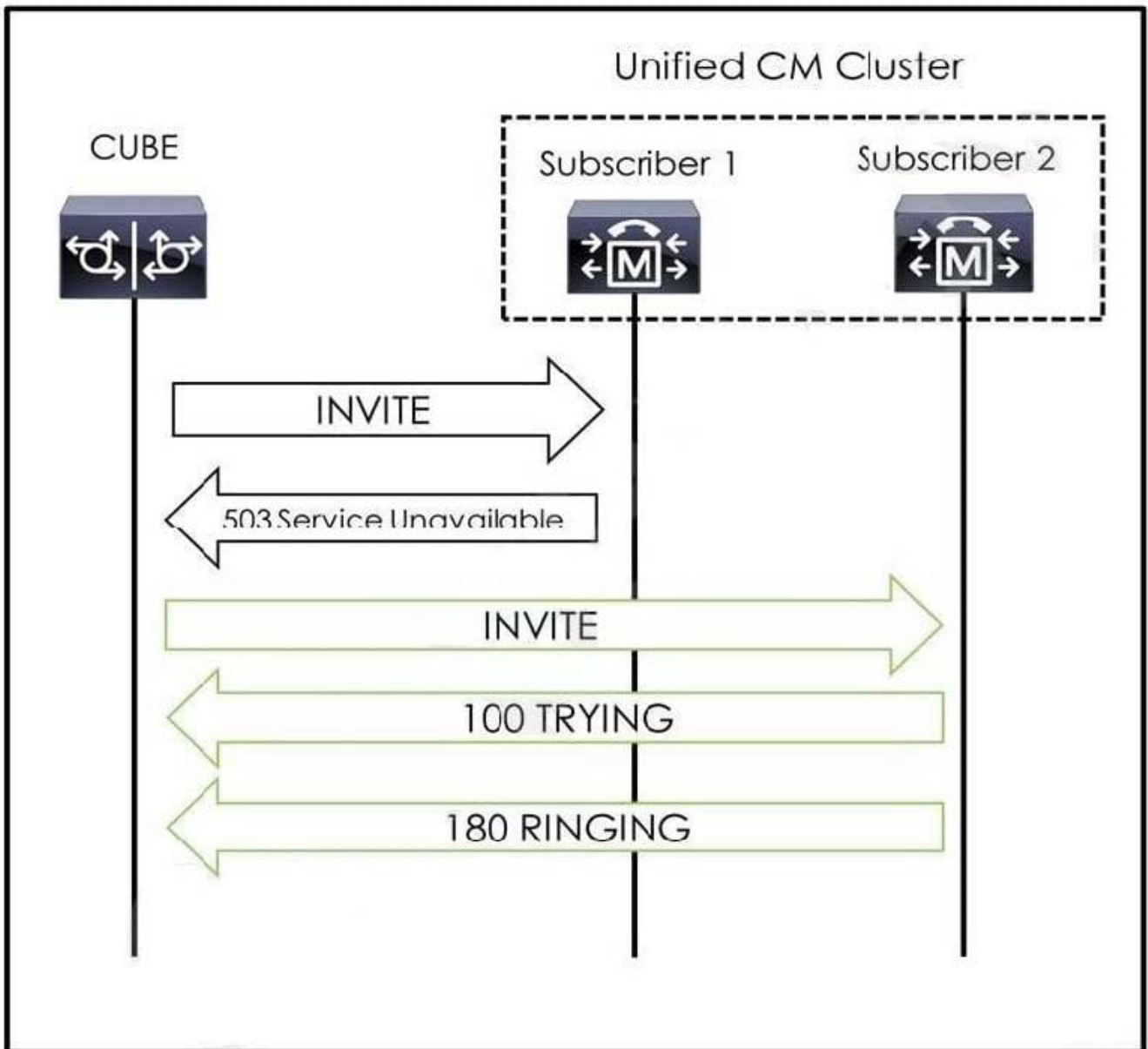
- A. Classify all route patterns as on-net and prohibit on-net to on-net call transfers in Cisco Unified CM service parameters.
 - B. Classify all route patterns as off-net and prohibit off-net to off-net call transfers in Cisco Unified CM service parameters.
 - C. Classify all route patterns as on-net or off-net and prohibit off-net to off-net call transfers in Cisco Unified CM service parameters.
 - D. Inbound CSS on any gateway typically should have access to internal destinations and PSTN destinations.
-

E. Inbound CSS on any gateway typically should have access to internal destinations only and not PSTN destinations.

Correct Answer: CE

QUESTION 11

Refer to the exhibit.



A customer submits this debug output, captured on a Cisco IOS router. Assuming that an MGCP gateway is configured with a ISDN BRI interface, which BRI changes resolve the issue?

SIP Trunk Security Profile Information

| | |
|--|---|
| Name* | <input type="text" value="CUP Non Secure SIP Profile"/> |
| Description | <input type="text"/> |
| Device Security Mode | <input type="text" value="Non Secure"/> |
| Incoming Transport Type* | <input type="text" value="TCP +UDP"/> |
| Outgoing Transport Type | <input type="text" value="TCP"/> |
| <input type="checkbox"/> Enable Digest Authentication | |
| Nonce Validity Time (mins)* | <input type="text" value="600"/> |
| Secure Certificate Subject or Subject Alternate Name | <input type="text"/> |
| Incoming Port* | <input type="text" value="5060"/> |
| <input type="checkbox"/> Enable Application level authorization | |
| <input checked="" type="checkbox"/> Accept presence subscription | |
| <input checked="" type="checkbox"/> Accept out-of-dialog refer** | |
| <input type="checkbox"/> Accept unsolicited notification | |
| <input type="checkbox"/> Accept replaces header | |
| <input type="checkbox"/> Transmit security status | |
| <input type="checkbox"/> Allow charging header | |
| | <input type="text" value="Use Default filter"/> |

SIP V.150 Outbound SDP Offer Filtering*

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

Correct Answer: C

QUESTION 12

What is a valid class included in the 8-Class QoS Strategy in a VoIP network?

- A. Assured Forwarding
- B. Broadcast Video
- C. Multimedia Conferencing
- D. Real-Time Interactive

Correct Answer: C

Reference: <https://www.ciscopress.com/articles/article.asp?p=2756478andseqNum=8>

QUESTION 13

Refer to the Exhibit.

```
Router# configure terminal
Router(config)# interface gigabitethernet 5/1
Router(config-if)# description Cube E41.228-0197
```

Which command is required to allow this media resource to handle Video Media streams?

- A. maximum sessions hardware 50
- B. video codec h264
- C. codec pass-through
- D. associate application Cisco Unified Border Element

Correct Answer: C

The command "codec pass-through" would let media streams be negotiated later in the call.

<https://community.cisco.com/t5/ip-telephony-and-phones/mtp-codec-pass-through/td-p/2159568>

QUESTION 14

What are two differences between media flow-around and media flow-through on cisco unified Border element?
(Choose two.)

- A. When using media flow-through, the call signaling and media are passed through the Cisco Unified Border Element
- B. When using media flow-through, the call signaling goes through the Cisco Unified Border Element, but media is not passed through it
- C. When using media flow-around. the call signaling goes through the Cisco Unified Border Element, but media is not passed through it
- D. When using media flow-around, both call signaling and media do not go through the Cisco Unified Border Element

E. When using media flow-through, call signaling goes through the Cisco Unified Border Element but media does not

Correct Answer: AC

Reference: <https://www.cisco.com/c/en/us/td/docs/ios-xml/ios/voice/cube/configuration/cube-book/media-path.html>

QUESTION 15

An engineer with ID012345678 must build an international dial plan in Cisco Unified Communications Manager. Which action should be taken when building a variable-length route pattern?

- A. reduce the T302 timer to less than 4 seconds
- B. configure single route pattern for international calls
- C. create a second route pattern followed by the # wildcard
- D. set up all international route patterns to 0.1

Correct Answer: C

When building a variable-length route pattern, you need to create a second route pattern followed by the # wildcard. This will allow the user to indicate the end of the number by dialing #. For example, if you want to create a route pattern for

international calls, you would create a route pattern like this:

9.011!#

This route pattern will match any number that starts with 9.011, followed by any number of digits, and then ends with #.

The other options are incorrect because:

Configuring a single route pattern for international calls will not allow the user to indicate the end of the number.

Setting up all international route patterns to 0.1 will not allow the user to indicate the end of the number.

Reducing the T302 timer to less than 4 seconds will not allow the user to indicate the end of the number.

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