

400-051^{Q&As}

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QUESTION 1

How many destinations can be configured for a SIP trunk on a Cisco Unified Communications Manager 9.1 system when the destination address is an SRV?

A. 1

- B. 2
- C. 3
- D. 8
- E. 16

Correct Answer: A

SIP trunks can be configured with up to 16 destination IP addresses, 16 fully qualified domain names, or a single DNS SRV entry.

References: http://www.cisco.com/c/en/us/td/docs/voice_ip_comm/cucm/srnd/8x/uc8x/trunks.html

QUESTION 2

Refer to the exhibit.



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Assume the B-ACD configuration on a Cisco IOS Cisco Unified Communications Manager Express router is operational. What will happen to a new call that enters the call queue when there are already two calls in queue?

- A. The call will be forwarded to extension 2120.
- B. The call will be forwarded to extension 2220.
- C. The call will be forwarded to extension 2003.
- D. The call will be disconnected with user busy.
- E. The call will be forwarded to 2100.
- Correct Answer: C

That is because queue over flow is forwarded to 2003 and maximum number of calls in queue is configured as two.



QUESTION 3

Refer to the exhibit.

```
[ContactServiceEdgeHandlerLogger] [ContactServiceEdgeHandler::edgeIsActive] -
[csf.httpclient] [http::CurlHttpUtils::logOperationTiming] - Network IO timestamps: [name
lookup = 0 ; connect = 0 ; ssl connect = 0 ; pre-transfer = 0 ; start-transfer = 0 ;
total = 0.203 ; redirect = 0]
[csf.httpclient] [http::CurlAnswerEvaluator::curlCodeToResult] - curlCode=[7] error
message=[Failed to connect to 172.16.100.51 port 7070; Connection refused]
result=[HOST_UNREACHABLE ERROR] fips enabled=[false]
[csf.httpclient] [http::/executeImpl] - *----* HCTP response from:
http://voicemailserver:7080/vmevents/cometd/handshake [18] -> 0.
[csf.httpclient] [http::executeImpl] - There was an issue performing the call to
curl_easy_perform: HOST_UNREACHABLE_ERROR <
[csf.httpclient] [http::HttpRequestData::returnEasyCURLConnection] - Returning borrowed
EasyCURLConnection from request : 📢 •
[csf.edge.capability.EdgeAccessDirector] [edge::EdgeAccessDirector::getInstance] -
Registering this as a DefaultReliciesStore observer
[csf.voicemail] [NotificationClient::sendRequest] - [this: 0D06A400] Http operation error
4 , HOST_UNREACHABLE_ERROR
```

A Jabber for Windows application fails to connect to the voicemail server. Which two options cause this problem? (Choose two)

- A. The jetty service has been disabled or is not running.
- B. A voicemail user password configuration error exists.
- C. A firewall is configured for blocking port 7080.
- D. An SSL certificate has an encryption problem.
- E. A company internal DNS server has a timeout problem.

Correct Answer: CE

QUESTION 4

A UCCX manager is monitoring several groups and has added a new team for the finance department. The manager can monitor all team members except those that have just been in the finance department. Which UCCX administration steps can resolve the issue?

- A. Subsystem>Team>Assign Supervisor and Contact Service Queue
- B. Wizards> RmCm Wizards>Modify Existing Service Queue
- C. Tools>User Management>Supervisor Capability View
- D. Subsystem>RmCm.>Resources
- E. Tools>User Management>Agent Capability View
- F. Subsystem>RmCm>Contact Service Queue



Correct Answer: D

QUESTION 5

Refer to the exhibit.

Received: INVITE sip:+15552341234@10.41.11.21:5060 SIP/2.0 Via: SIP/2.0/TCP 10.41.11.10:5060; br anch=z9hG4bK4bb12565f6120d Via: SIP/2.0/TCP 10.10.20.20:5060; br anch=z9hG4bK2d4790; received=192.0.2.111 Via: SIP/2.0/TCP 10.50.40.30:5060; br anch=z9hG4bK2d6014; received=192.0.2.20 From: "Test" <sip:5559874321@customer.com>; tag=15269661~58798a82-6f11 4d0c-9024-092b2761f953-71117 To: <sip:+15552341234@10.41.11.21> Cisco-Guid: 2960899712-000065536-0000112086-0202057994 P-Asserted-Identity: "Test" <sip:5559874321@customer.com>; party-alling; screen=yes; privacy=off Contact: <sip:5559874321@10.41.11.10:5060; transport="tcp"></sip:5559874321@10.41.11.10:5060;></sip:5559874321@customer.com></sip:+15552341234@10.41.11.21></sip:5559874321@customer.com>	600
voice translation-rule 1 rule 1 /\+/ // voice translation-rule 2 rule 1 /\+1/ // voice translation-rule 3	
rule 1 /\+1/ /91/ voice translation-profile 1 translate called 1	
voice translation-profile 2 translate called 2 voice translation-profile 3 translate called 3	

A call is received on a Cisco Unified Border Element gateway. The debug captures this SIP INVITE message. Which configurations result in 915556781234 as the final called number?

A. voice class uri 1 siphost ipv4:10.41.11.10!voice class uri 2 siphost ipv4:10.50.40.30!dial-peer voice 1 voipincoming uri via 1translation-profile incoming 1!dial-peer voice 2 voipincoming uri via 2translation-profile incoming 2!dial-peer voice 3 voipincoming called-number +1Ttranslation-profile incoming 3

B. voice class uri 1 siphost dns:customer.comvoice class uri 2 siphost ipv4:10.50.40.30dial-peer voice 1 voipincoming uri via 1translation-profile incoming 1dial-peer voice 2 voipincoming uri via 2translation-profile incoming 2dial-peer voice 3 voipincoming called-number +1Ttranslation-profile incoming 3

C. voice class uri 1 siphost ipv4:10.41.11.10voice class uri 2 siphost ipv4:10.50.40.30dial-peer voice 1 voipincoming called-number +1[2-9]..[2-9].....translation-profile incoming 3dial-peer voice 2 voipincoming uri via 2translation-profile incoming 2dial-peer voice 3 voipincoming called-number [2-9]..[2-9].....translation-profile incoming 1

D. voice class uri 1 siphost dns:customer.comvoice class uri 2 siphost ipv4:10.50.40.30dial-peer voice 1 voipincoming uri via 1translation-profile incoming 1dial-peer voice 2 voipincoming uri via 2translation-profile incoming 1dial-peer voice 3 voipincoming called-number +1Ttranslation-profile incoming 3

E. voice class uri 1 siphost ipv4:10.41.11.10voice class uri 2 siphost ipv4:10.50.40.30voice class uri 3 siphost dns:customer.comdial-peer voice 1 voipincoming uri via 1translation-profile incoming 2dial-peer voice 2 voipincoming uri via 2translation-profile incoming 1dial-peer voice 3 voipincoming uri via 3incoming called-number +1Ttranslation-profile incoming 1

F. voice class uri 1 siphost ipv4:10.41.11.10voice class uri 2 siphost ipv4:10.50.40.30voice class uri 3 siphost dns:customer.comdial-peer voice 1 voipincoming uri via 1translation-profile outgoing 3dial-peer voice 2 voipincoming uri via 2translation-profile incoming 2dial-peer voice 3 voipincoming called-number +1Ttranslation-profile incoming 2



Correct Answer: B

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