



# 300-815<sup>Q&As</sup>

Implementing Cisco Advanced Call Control and Mobility Services  
(CLACCM)

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**QUESTION 1**

Refer to the exhibit.

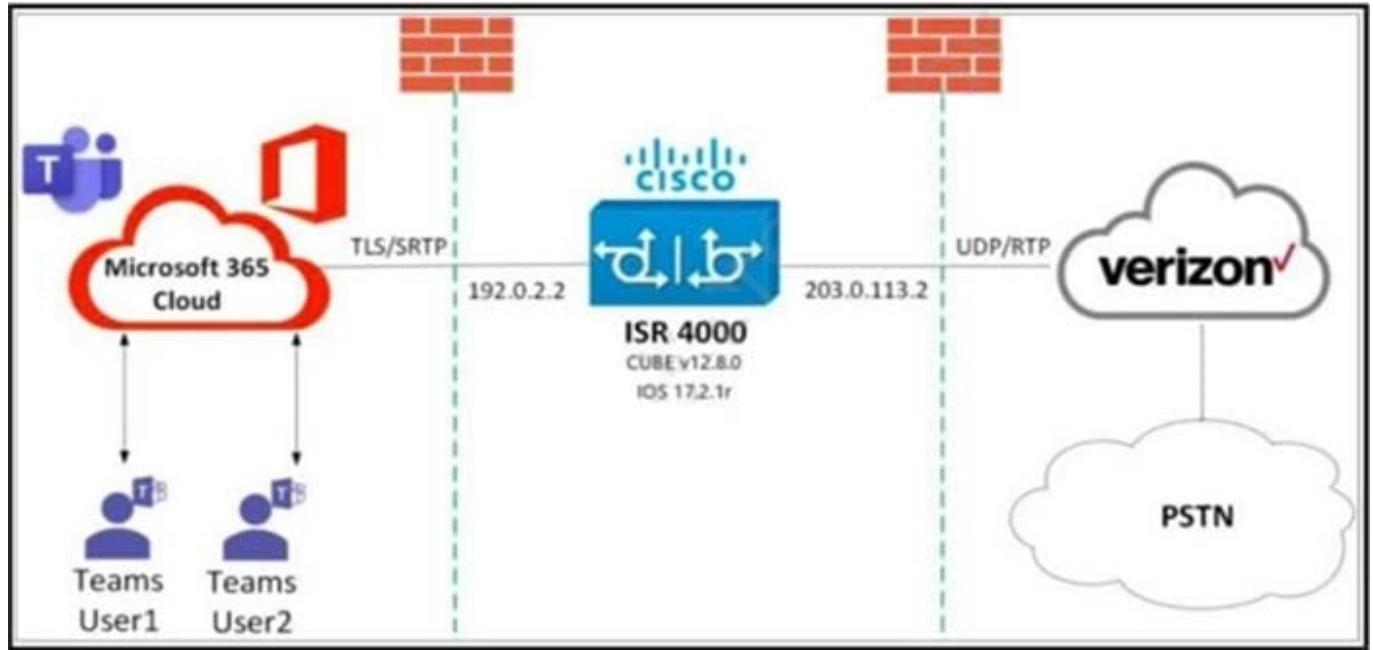
```
SIPHandler/ccbId=0/scbId=0/wait_SIPTimer: TimerExpired
type=SIP_TIMER_WAIT_CONNECT value=5000 retries=0
Stack/Transport/0x0xee9c8980/sipTransportPostInternalMsg: Posting Internal Msg
type=1
Stack/Transport/0x0/sipTransportPostCloseConnection: Posting TCP conn close for
addr=10.10.5.11, port=5060, connid=20
Stack/Transport/0x0/sipDeleteConnInstance: Deleted conn=0xe7ac06c0, connid=20,
addr=10.10.5.11, port=5060, transport=ICP
Stack/Info/0x0/ccsip_process_sipsip_queue_event: ccsip_spi_get_msg_type returned:
2 (SIP_NETWORK_MSG), for event 64 (SIPSPI_EV_INTERNAL_MSG)
Stack/Error/0x0xee9c8950/sipTransportPostSendFailure: Posting send failure msg
with tcb: (nil) reason=4
Stack/Info/0x0/ccsip_process_sipsip_queue_event: ccsip_spi_get_msg_type
returned: 2 (SIP_NETWORK_MSG), for event 55 (SIPSPI_EV_SEND_FAILURE_MSG)
Stack/Info/0x0xee9c8980/ccsip_spi_process_event: Send Error for event
(0xee9cb8b0)
Stack/Error/0x0/act_idle_send_msg_failure: Send Error to 10.10.5.11:5060 for
transport TCP
Stack/Info/0x0xee9c8980/ccsip_set_oo_cause_for_spi_err: Categorized cause: 38,
category:106
Stack/Info/0x0xee9c8980/sipSPIInitiateDisconnect: Initiate call disconnect (38)
for outgoing call
SIPHandler/ccbId=22609/scbId=0/ccsip_api_call_disconnected: ccb->cc_disc_cause
(38): ccb->sip_disc_cause (503)
SIPHandler/ccbId=22609/scbId=0/findDevicePID: Routed to SIPD by ccbId/scbId
Stack/States/0x0xee9c8980/sipSPIChangeState: 0xee9c8980 : State change from
(STATE_IDLE, SUBSTATE_NONE) to (STATE_DISCONNECTING, SUBSTATE_NONE)
```

An administrator has configured a SIP trunk between two Cisco UCM clusters. For calls that should use the trunk, the calls fail with a fast busy. The administrator checks the Cisco CallManager SDL traces and found that the cluster to which the calling device is registered never sends an INVITE to the destination cluster. The administrator also verifies that all nodes from both clusters are powered on, and the CallManager service is running. How is this issue resolved?

- A. The administrator must associate the route pattern with a calling search space the device can dial.
- B. The administrator needs to enable OPTIONS pings on the SIP trunks for both clusters.
- C. The administrator must allow connectivity so that TCP connections do not fail between the nodes.
- D. The administrator needs to disable OPTIONS pings on the SIP trunks for both clusters.

Correct Answer: C

**QUESTION 2**



Refer to the exhibit. A company is using Microsoft Teams with Cisco Unified Border Element integration, but the administrator sees a one-way audio issue with Microsoft Teams. The administrator must modify the SIP profile to send the proper information on the SDP for IP address for media to match the internal and external interface. Which set of commands resolves the issue?

- A. voice class sip-profiles 1 request INVITE sdp-header Connection-Info modify "2\76\1" "2.78.1"
- B. voice class sip-profiles 1 request INVITE sdp-header Session-Owner modify "27\0\0" "27.3.0"
- C. voice class sip-profiles 1 request INVITE sdp-header Connection add "2\76\1" "2.78.1"
- D. voice class sip-profiles 1 request INVITE sdp-header Audio-Attribute modify "2\76\1" "2.71.1"

Correct Answer: C

### QUESTION 3

Customers that call into a company's IVR report that when they try to select an option, none of the prompts work. The administrator determines that the calls are coming in across an H.323 gateway. While analyzing the dial peer that points toward Cisco UCM, the administrator notices that no DTMF method is configured. Which command resolves this issue?

- A. dial-peer voice 2 voip dtmf-relay sip-kpml
- B. dial-peer voice 2 voip dtmf-relay h245-alphanumeric
- C. dial-peer voice 2 voip dtmf-relay sip-notify
- D. dial-peer voice 2 pots dtmf-relay h245-alphanumeric

Correct Answer: B



#### QUESTION 4

A network engineer designs a new dial plan and wants to block a certain range of numbers (8135100 through 8135105). Which route pattern should be configured to block only the numbers in this range?

- A. 813510[012345]
- B. 813510[12345]
- C. 813510[^0-5] D. 81XXXXX

Correct Answer: A

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#### QUESTION 5

An engineer has a hunt group with some overflow and many calls are going to voicemail. To reduce the number of calls forwarded to voicemail, the engineer must create Call Queuing to make callers wait before going to voicemail. How will Call Queuing affect Forward Hunt No Answer and Forward Hunt Busy?

- A. Forward Hunt No Answer and Forward Hunt Busy must be enabled while Call Queuing is enabled.
- B. After Call Queuing is enabled, Forward Hunt No Answer and Forward Hunt Busy must be disabled manually.
- C. Before Call Queuing is enabled, Forward Hunt No Answer and Forward Hunt Busy must be disabled manually.
- D. When Call Queuing is enabled, Forward Hunt No Answer and Forward Hunt Busy are disabled automatically.

Correct Answer: D

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