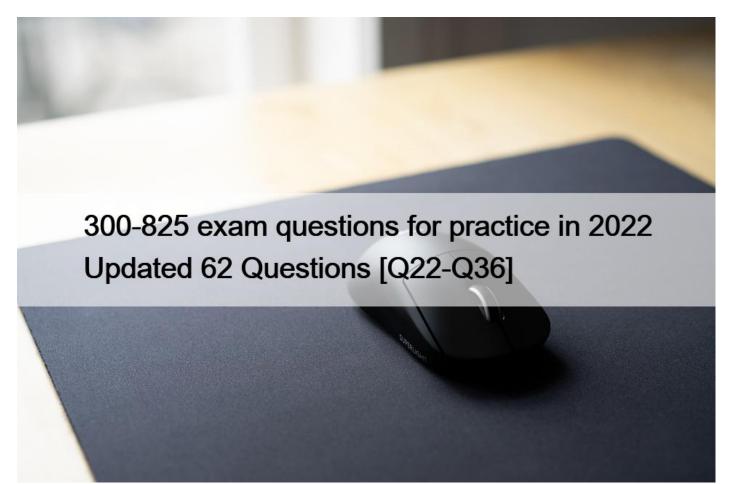
# 300-825 exam questions for practice in 2022 Updated 62 Questions [Q22-Q36



300-825 exam questions for practice in 2022 Updated 62 Questions Updated Jun-2022 Premium 300-825 Exam Engine pdf - Download Free Updated 62 Questions

NO.22 What is the maximum number of full HD external web app calls that is supported per large OVA Expressway?

- \* 800
- \* 600
- \* 150
- \* 200

**NO.23** An administrator notices that the Cluster section is not available under the configuration menu in the Cisco Meeting Server web interface. There are three Cisco Meeting Server servers that must run Call Bridge and database services in a single cluster. Which configuration should be completed?

- \* Certificates must be configured on the Call Bridge.
- \* Call Bridge groups must be created in the API.
- \* Call Bridges must be connected to a database cluster.
- \* Call Bridge clustering must be configured in the CLI.

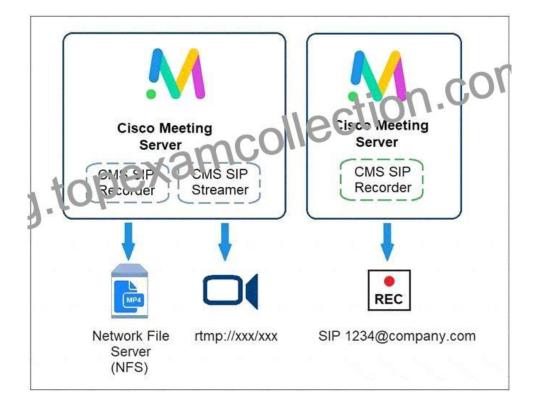
NO.24 After performing an upgrade of the Cisco Meeting Server cluster, an engineer sees an error message indicating that Cisco

Meeting Server is not licensed. Also, no spaces or users show in the GUI. What should be done to resolve this issue?

- \* The database cluster upgrade\_schema command must be run on the new database cluster master.
- \* The Cisco Meeting Server nodes must be re-upgraded using the same file to upgrade all nodes.
- \* The Cisco Meeting Server nodes must be re-upgraded with the database master being first and then all peers.
- \* New database certificates must be generated for each peer in the cluster.

Explanation/Reference: https://www.cisco.com/c/dam/en/us/td/docs/conferencing/ciscoMeetingServer/Release\_Notes/ Version-3-0/Cisco-Meeting-Server-Release-Notes-3-0.pdf

**NO.25** Refer to the exhibit.



Which two features are supported in the Cisco Meeting Server recording and streaming solution? (Choose two.)

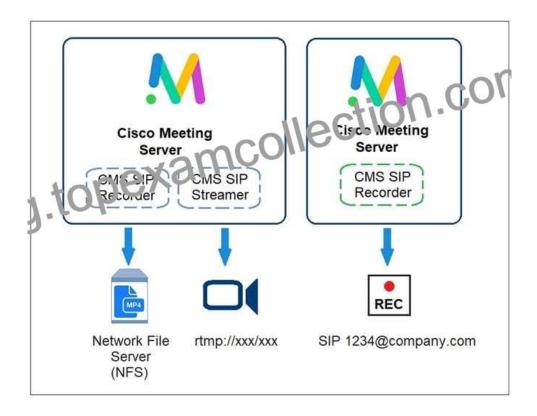
- \* editing recordings
- \* portal to access recordings
- \* 1080p resolution for streaming
- \* changing layouts
- \* closed captioning in streaming

NO.26 How should an administrator configure Skype for Business to work with a Cisco Meeting Server cluster?

- \* with the CsTrustedApplicationPool on the Lync/Skype server to specify the Cisco Meeting Server Web Admin FQDN
- \* with the CsTrustedApplication Pool on the Lync/Skype server to specify port 5060
- \* with CsStaticRoute on the Lync/Skype server to the first Call Bridge in the cluster
- \* with the CsTrustedApplication Pool ComputerFqdn to the first Call Bridge in the cluster

Explanation/Reference: https://www.cisco.com/c/en/us/support/docs/conferencing/meeting-server/212217-configure-cisco-meeting-server-and-skype.html#anc13

NO.27



Refer to the exhibit. Which two features are supported in the Cisco Meeting Server recording and streaming solution? (Choose two.)

- \* editing recordings
- \* portal to access recordings
- \* 1080p resolution for streaming
- \* changing layouts
- \* closed captioning in streaming

Explanation/Reference: https://www.cisco.com/c/dam/en/us/td/docs/solutions/PA/mcp/DEPLOYMENT\_MEETINGS-CWMS\_to\_CMS.pdf

NO.28 In the Cisco TMS GUI, a Cisco UCM registered system shows these error details:

- \* Provisioning mode is set to Cisco UCM for this system.
- \* The system cannot be refreshed or scheduled.
- \* To manage this system as a Cisco TMS system, the provisioning mode needs to be changed to Cisco TMS.
- \* To add the system as Cisco UCM managed system, it must be added through the Cisco UCM that is provisioning the system.

What should be done to resolve this error?

- \* Purge the system in Cisco TMS and add the system back directly to Cisco TMS using "Add by IP Address".
- \* Restart Microsoft IIS on the Windows server that hosts Cisco TMS and perform a "Force Refresh" on the system.
- \* Delete the system and add the system back with the "Add from Cisco UCM or Cisco TMS" option.
- \* Update the credentials for the system in Cisco TMS.

# NO.29 Refer to the exhibit.

guest login request 562763450: resolution in progress

guest login request 562763450: call ID lookup scheduled

guest login request 562763450: resolution in progress

guest login request 562763450: credential storage scheduled (queue length: 1) created guest account with user ID "quest487660227" guest login request 562763450: credential storage executed guest login request 562763450: credential storage in progress guest login request 562763450: successfully stored credentials instantiating user "guest487660227" conference f5771a88-0f36-4b58-8dbd-934aaa579ebf: lock et al. account with user ID conference f5771a88-0f36-4b58-8dbd-934aaa579ebf: lock accessfully stored credentials instantiating user conference f5771a88-0f36-4b58-8dbd-934aaa579ebf: lock state has changed to locked API "3.0 Record Test Space" Space GUID: 71b4f99f-2365-4730-b3d9-d2d61505150c <&#8211;> Call Correlator GUID: 334b0c10-6f85-4760-97ab- conference f5771a88-0f36-4b58-8dbd-934aaa579ebf: lock state has changed to unlocked starting automatic recording (space &#8220;3.0 Record Test Space&#8221;) API call leg 121d2c87-6940-4483-8d3d-71ff162947ba in call f5771a88-0f36-4b58b8dbd-934aaa579ebf (API call 5a0a7f49-78f8-46fd-add9-4a conference f5771a88-0f36-4b58-8dbd-934aaa579ebf has control/media GUID:

#### d7f2d940-c7d9-4421-8f13-7bf94756b08a

conference f5771a88-0f36-4b58-8dbd-934aaa579ebf named "3.0 Record Test Space" unable to start recording – recorder URI not configured new session created for user "guest487660227" call 49: allocated for guest487660227 "Web" conference participation call 49: configured – API call leg 121d2c87-6940-4483-8d3d-71ff162947ba call 49: setting up combined RTP session for DTLS (combined media and control) participant "quest487660227" joined space 71b4f99f-2365-4730-b3d9-d2d61505150c (3.0 Record Test Space) participant "quest487660227" (121d2c87-6940-4483-8d3d-71ff162947ba) joined conference f5771a88-0f36-4b58-8dbd-934aaa579ebf via W call 49: starting DTLS combined media negotiation (as initiator) call 49: completed DTLS combined media negotiation invalid API operation – unhandleable URI /api/v1/recorders user "guest487660227" conference media) call 449: destroying API call leg 121d2c87-6940-4483-8d3d-71ff162947ba An administrator tries to implement a recording with a Cisco Meeting Server version 3.0 deployment. A separate VM is deployed and configured with the recorder service. During testing, it is discovered that the recording is failing, and the administrator collects logs.

Which configuration step should be completed?

- \* In the coSpace configuration, a secondaryUri must be configured.
- \* In the callLegProfile, recordingControlAllowed must be set to true.
- \* In the Outbound dial rules, the recording domain must have a rule.
- \* In the CallProfile, the sipRecorderUri must be configured.

NO.30 What is one of the ways Cisco TMS redundancy is configured?

- \* Set up three Cisco TMS servers by using F5 BIG-IP for load balancing IP traffic.
- \* Set up one Cisco TMS server with an SQL database for automatic failover.
- \* Set up one Cisco TMS server by using a Hot Standby model.
- \* Set up two Cisco TMS servers with a network load balancer in front.

**NO.31** An administrator sets up conference resources by using a clustered Cisco Meeting Server deployment for video endpoints that are registered to Expressway. The administrator must ensure that no single server gets overloaded within the cluster. What should be configured to accomplish this goal?

\* a record with the cluster FQDN of the Cisco Meeting Server in the external DNS servers.

- \* Expressway-C as the traversal client to Expressway-E.
- \* Cisco Meeting Server and Expressway-C in the same domain.
- \* a separate zone for Cisco Meeting Server servers in the cluster.

**NO.32** An administrator configures a Call Bridge to record meetings. Which two configuration elements tell the Call Bridge where to find the recording server? (Choose two.)

- \* tenants
- \* IVR
- \* DTMF combination
- \* sipRecorderUri
- \* outbound rules

Explanation/Reference: https://www.cisco.com/c/dam/en/us/td/docs/conferencing/ciscoMeetingServer/Deployment\_Guide/ Version-3-0/Cisco-Meeting-Server-3-0-Single-Combined-Server-Deployment.pdf

**NO.33** An engineer configures Cisco Meeting Server with Expressway for external web app access and must enable participants to use the web app to join conferences that are hosted on the meeting server. Which service should be used in the Expressway-E accomplish this goal?

- \* Cisco Meeting Server
- \* SIP
- \* XMPP server
- \* TURN

Explanation/Reference: https://www.cisco.com/c/dam/en/us/td/docs/conferencing/ciscoMeetingServer/Installation/Cisco-Meeting-Server-Deployment-Planning-and-Preparation-with-Expressway-Guide.pdf

**NO.34** An administrator configures a Call Bridge to record meetings. Which two configuration elements tell the Call Bridge where to find the recording server? (Choose two.)

- \* tenants
- \* IVR
- \* DTMF combination
- \* sipRecorderUri
- \* outbound rules

NO.35 Which component does the Cisco Meeting Server web app use for firewall traversal?

- \* XMPP server
- \* MMP interface
- \* TURN server
- \* Call Bridge

Explanation/Reference: https://www.cisco.com/c/dam/en/us/td/docs/conferencing/ciscoMeetingServer/Deployment\_Guide/ Version-2-9/Cisco-Meeting-Server-2-9-Single-Split-Server-Deployment.pdf

NO.36 outgoing SIP TLS data to 10.0.101.111:5061 from 10.0.101.121:43332, size 5430:

# BEGINNING OF MESSAGE

# INVITE sip:jabberuser@pod1.local SIP/2.0

Via: SIP/2.0/TLS 10.0.101.121:5061;branch=z9hG4bK6e7932e8ae287a80ee699eb04a6155a2 Call-ID:

cf600c54-974c-48e5-92e2-95b1d4b9218f CSeq: 1199885731 INVITE Max-Forwards: 70 Contact:

<sip:8001@10.0.101.121;transport=tls>;isFocus To: <sip:jabberuser@pod1.local> From: &#8220;CCNP Collaboration&#8221; <sip:8001@pod1.local>;tag=73a834a850d8ed13 Replaces:

5e158900-10001-ee62f-0@10.0.101.111;to-tag=1769141~ba2cc769-5032-4f8f- a864-9a51 Allow: INVITE, ACK, CANCEL, OPTIONS, INFO, BYE, UPDATE, REFER, SUBSCRIBE, NOTIFY, MESSAGE Supported: timer, X-cisco-callinfo Session-Expires: 1800 Min-SE: 90 User-Agent: Acano CallBridge

# END OF MESSAGE

incoming SIP TLS data from 10.0.101.111:5061 to 10.0.101.121:43332, size 402:

#### BEGINNING OF MESSAGE

#### SIP/2.0 403 Forbidden

Via: SIP/2.0/TLS 10.0.101.121:5061;branch=z9hG4bK6e7932e8ae287a80ee699eb04a6155a2 From: "CCNP Collaboration" <sip:8001@pod1.local>;tag=73a834a850d8ed13 To: <sip:jabberuser@pod1.local>;tag=252834177 Date: Mon, 30 Nov 2020 18:25:12 GMT Call-ID: cf600c54-974c-48e5-92e2-95b1d4b9218f CSeq: 1199885731 INVITE Allow-Events: presence Server: Cisco-CUCM12.5 Content-Length: 0 END OF MESSAGE Refer to the exhibit. Some calls fail to connect after an engineer implements load balancing for a Cisco Meeting Server cluster. The engineer determines that if a call leaves Cisco UCM using the Cisco Meeting Server 1 SIP trunk, but the conference is already hosted on Cisco Meeting Server 2, then the call fails. If the call leaves Cisco UCM using the Cisco Meeting Server 2 SIP trunk, but the conference is already hosted on Cisco Meeting Server 1, then the call fails. What is the cause of the issue?

- \* Cisco UCM is configured with the wrong X.509 certificate name for the Call Bridge.
- \* Cisco UCM is sending calls to a Cisco Meeting Server when the calls are hosted on another Call Bridge in the same cluster.
- \* The Cisco UCM SIP trunk security profile is not configured to accept the replaces header.
- \* Cisco UCM does not have a SIP trunk to the Call Bridge.

Authentic 300-825 Dumps With 100% Passing Rate Practice Tests Dumps: https://www.topexamcollection.com/300-825-vce-collection.html]