

300-815 Test Valid & 300-815 Exam Question

```
voice translation-profile incoming
  translate called 999
!
voice translation-rule 999
  rule 1/\ (^([1-2] [1-2] [1-2]) 333\ ([4-5] [4-5] .\ ) $ / / \2333\1/
!
dial-peer voice 999 voip
  translation-profile outgoing incoming
  session protocol sipv2
  incoming called-number
  dtmf-relay rtp-nte
  codec transparent
  destination dpg 888
  no vad
!
voice class dpg 888
  dial-peer 999
!
dial-peer voice 888 voip
  destination-pattern 888
  session protocol sipv2
  session target ipv4:192.168.0.1
  codec transparent
  dtmf-relay rtp-nte
  no vad
```

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Cisco 300-815 Exam is designed for professionals who want to validate their knowledge and expertise in implementing Cisco Advanced Call Control and Mobility Services. Implementing Cisco Advanced Call Control and Mobility Services certification exam is part of the Cisco Certified Network Professional (CCNP) Collaboration program and is intended for individuals who are responsible for designing, configuring, and deploying advanced collaboration solutions in enterprise environments.

>> 300-815 Test Valid <<

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Cisco Implementing Cisco Advanced Call Control and Mobility Services Sample Questions (Q70-Q75):

NEW QUESTION # 70

Which IOS command creates a SIP-enabled dial peer?

- A. dial-peer voice 20 pots
- B. dial peer voice 20 sip
- C. dial-peer voice 20 voip
- D. voice dial-peer 20 sip

Answer: C

NEW QUESTION # 71

Which description of RTP timestamps or sequence numbers is true?

- A. Sequence numbers increase by four for each RTP packet transmitted.
- **B. The timestamp is used to place the incoming audio and video packets in the correct timing order (playout**
- C. The sequence number is used to detect losses.
- D. Timestamps increase by the time "carrying" by a packet.

Answer: B

Explanation:
delay compensation).

NEW QUESTION # 72

Refer to the exhibit.

Intercluster Lookup Service Configuration

Role: Hub Cluster

[Register to Another Hub...](#)

Exchange Global Dial Plan Replication Data with Remote Clusters

Advertised Route String *: CCNP

Synchronize Clusters Every *: 10 (1-1440 minutes)

ILS Authentication

Use TLS Certificates

Use Password

Password *:

Confirm Password *:

ILS Clusters and Global Dial Plan Imported Catalogs

Cluster ID/Name	Last Contact Time	Role	Advertised Route String	USN Data Synchronization Status
StandAloneCluster	2/17/21 10:31 AM	Hub	CCIE	Not Applicable
StandAloneCluster -		Hub (Local Cluster)	CCNP	Disabled

ILS has been configured between two hubs using this configuration. The hubs appear to register successfully, but ILS is not functioning as expected. Which configuration step is missing?

- A. A password has never been set for ILS.
- B. Use TLS Certificates must be selected.
- C. Trust certificates for ILS have not been installed on the clusters
- **D. The Cluster IDs have not been set to unique values**

Answer: D

NEW QUESTION # 73

Which two configuration parameters are prerequisites to set Native Call Queuing on Cisco Unified Communications Manager? (Choose two.)

- **A. Cisco RIS data collector service must be running on the same server as the Cisco CallManager service.**
- B. The phone button template must have the Queue Status Softkey configured.
- C. The maximum number of callers allowed in queue must be 10.
- D. A unicast music on hold audio source must be configured.
- **E. Cisco IP Voice Media Streaming Service must be activated on at least one node in the cluster.**

Answer: A,E

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