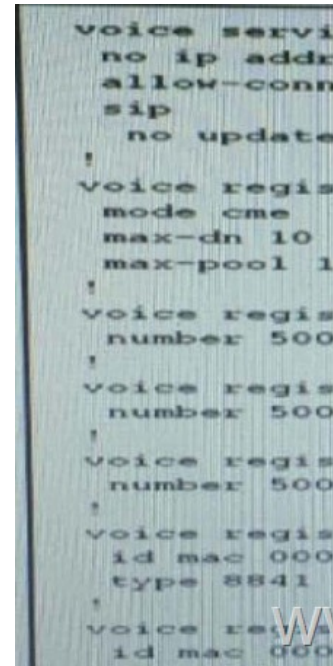


## [2016-New Lead2pass Cisco 400-051 VCE And PDF Instant Download (396-406)]

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QUESTION 396 Refer to the exhibit. An engineer is trying to provision CUCME with three 8841 phones. However all phone fail to register. Which two changes in the configuration would allow the phones to register? (Choose two)



```
voice service voip
  no ip address
  allow-connections sip
  sip
  no update-contacts
  !
voice register global
  mode cme
  max-dn 10
  max-pool 1
  !
voice register voip
  number 500
  !
voice register voip
  number 500
  !
voice register voip
  number 500
  !
voice register voip
  id mac 000
  type 8841
  !
voice register voip
  id mac 000
```

A. The registrar server command must be added under the voice register global configuration

B. The IP address trusted authenticate command must be added under voice service voip

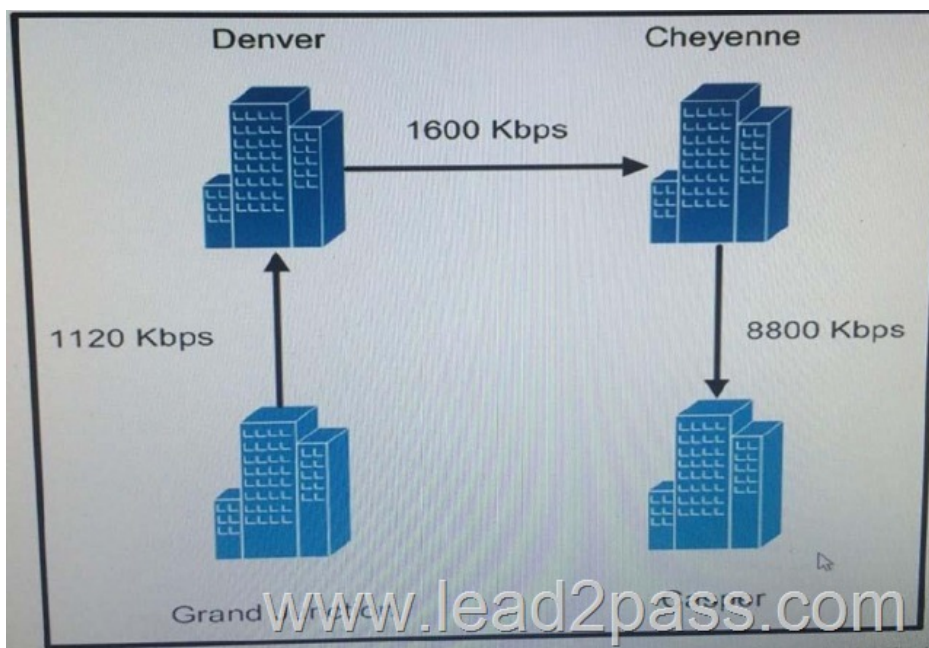
C. The source-address command must be added under the voice register global configuration

D. The local SIP proxy address must be configuration under the sip-ua configuration

E. The registrar server command must be added under the sip section of voice service voip Answer: CE

QUESTION 397 A collaboration engineer has been asked to implement secure real-time protocol between a Cisco Unified CM and SIP gateway. Which option is a consideration for this implementation? A. Only T.38 and Cisco fax protocol are supported B. SIP require the all the time be sent in GMT C. Call hold RE-INVITE is not supported D. SRTP is supported only in cisco IOS 15.x and higher Answer: B

QUESTION 398 Refer to the exhibit. A collaboration engineer configures Cisco Unified CM location using G.711 and iLBC for each site. The bandwidth for each link is shown. Which two options represent the maximum concurrent number of calls supported from Grand Junction to Casper for each Codec? (Choose two)



A. 20 G.711 calls B. 18 G.711 calls C. 36 iLBC calls  
D. 42 iLBC calls E. 11 G.711 calls F. 51 iLBC calls

Answer: CE QUESTION 399 A collaboration engineer is troubleshooting an MOH problem on a Cisco IOS SIP gateway. While searching through a debug ccsp message output, which three parameters in the SIP messages can be used to determine if the call was placed on hold? (Choose three) A. OPTIONS WITH 301 CALLHOLD B. INVITE WITH a=INACTIVE C. INVITE WITH a=SENDONLY D. OPTION WITH c=INACTIVE E. c=IN IP4 0.0.0.0 F. BYE WITH A = CALLHOLD

Answer: BCE QUESTION 400 Refer to the exhibit. A cisco collaboration engineer discovers that an instance of IOS media termination point (MTP) could not maintain stable registration with CUCM. Call manager traces is showing in the exhibit. What is the reason for the flapping registration?

```
!CUCM_CALLMANAGER-6-StationConnectionError: 4[DeviceName=MTPSITEA][ReasonCode=4][ClusterID=StandAloneCluster]
[ModelID=ULFCUCMFCB10]: Station device is closing the connection
!AppInfo [New connection accepted, DeviceName=, TCPPid = [1.100.14.145],
[Pass=172.38.140.1, Port=8804, Device Controller=[0,0,0]
!SdLog [StationClose [Waiting [MediaTerminationPointControl(1,100,137,136) [StationInit(1,100,42,1)
[1,100,14,145,2*172.38.140.1*
[[RtV=0,0,0,0,0,0,0,0,0,0] CloseStationReason = 4 StationId =
!AppInfo [MediaTerminationPointControl(136) :!STAR_StationClose DeviceName= MTPSITEA
!AppInfo [MediaTerminationPointControl(136) :!devTotalCounter Count=100
!AppInfo [MediaTerminationPointControl(136) :!devAvailableCounter - Count=100
!AppInfo [MediaTerminationPointControl(136) :!devActiveCounter - Count=0
!AppInfo [MediaTerminationPointControl(136) :!devStateInUse - Count=
!Delete State Instance For MtpControl [
!SdLog [MediaTerminationPointControl(1,100,137,136) [StationInit(1,100,42,1)
!MediaTerminationPointControl(1,100,137,136) [StationInit(1,100,42,1)
!AppInfo [MediaTerminationPointControl(136) :!nulling_cdm_MediaTerminationPointControl - Device = MTPSITEA - Un-Registered
```

A. The CCM version on IOS configuration does not match the CUCM version. B. The IOS MTP is experiencing high CPU and is missing its keep-alive. C. A Firewall is blocking port 2000 intermittently between IOS Device and CUCM. D. Another IOS Media device is attempting to register with the same name.

Answer: D QUESTION 401 A collaboration engineer is designing Cisco IM&P implementation to support instant messaging logging for compliance. Which two external databases can be used to support that functionality? (Choose two.)

A. Oracle database B. MySQL database C. Microsoft SQL database D. PostgreSQL database E. Informix SQL database Answer: AD

QUESTION 402 Refer to the exhibit. A cisco collaboration engineer is troubleshooting a gateway and gatekeeper problem and sees this output from a debug command. Which two configuration can cause this problem? (Choose two)

```
RAS INCOMING PDU ::=
value RasMessage ::= registrationReject :
{
  requestSeqNum 24
  protocolIdentifier { 0 0 8 2250 0 3 }
  rejectReason duplicateAlias:
  {
  }
  gatekeeperIdentifier {"gk"}
}
```

A. The same zone prefix is configured in two different gatekeepers  
B. The same H323-ID is configured in two different gateways  
C. The same gw-type-prefix is configured in two different zone subnets IDs  
D. The same zone subnet ID is configured in two different gatekeepers  
E. The same E164-ID is configured in two different gateways  
Answer: BE QUESTION 403 The Cisco Unified Border Element is configured using high availability with the Hot Standby Routing Protocol. Which two pieces of information can be gathered about the calls traversing these border elements? (Choose two.)

```
CUBE_PSTN#show voice high

===== Voice HA DB INFO
Number of calls in HA DB:
Number of calls in HA syn
Number of calls in HA pre
```

A. The total number of calls is 150.  
B. The number of non-native calls is 70.  
C. The number of native calls is 50.  
D. The number of calls preserved is 220.  
E. The total number of active calls is 100.  
Answer: AB QUESTION 404 Refer to the exhibit. Which two SIP packet handling behaviour will result with this cisco Unified Border Element (CUBE) configuration? (Choose two)

```
voice service
no ip address
rtcp keepalive
mode border-e
allow-connect
redundancy
no supplement
signaling for
fax protocol
sip
min-se 360
header-pass
error-passthru
asserted-id
midcall-sign
privacy-pol
pass-thru h
pass-thru c
no call ser
```

A. Unsupported content/MIME pass-through  
B. SIP Refer is not support when received on this CUBE  
C. Privacy headers received on SIP message will be replaced with NON-privacy headers on this CUBE  
D. P-Preferred identities  
E. Mid-call codec changes  
Answer: AE QUESTION 405 A CUCM engineer has deployed Type B SIP Phones on a remote site and no SIP dial rules were deployed for these phones. How Will CUCM receive the DTMF after the phone goes off-hook and the button are pressed?  
A. Each digit will be received by CUCM in a SIP NOTIFY message as soon as they are pressed  
B. The first digit will be received in a sip invite and subsequent digits will be received using NOTIFY message as soon as they are pressed.  
C. Each digit will be received by CUCM in a SIP INVITE as soon as the dial soft key has been pressed.  
D. All digits will be received by CUCM in a SIP INVITE as soon as the dial soft key has been pressed  
Answer: A QUESTION 406 The Video engineer wants to enable the LATM codec to allow video endpoint to communicate over audio With other IP devices Which two Characteristic should the voice engineer be aware of before

enabling ATM on the Cisco Unified border element router? (Choose two) A. Dual tone Multi-frequency interworking with ATM codec is not supported B. Codec transcoding between ATM and other codecs is not supported C. SIP UPDATE message outlined in RFC3311 is not supported D. Box-to-Box High availability support feature is not supported E. Configure ATM under a voice class or dial peer is not supported F. Basic calls using flow-around or flow-through is not supported Answer: AB

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