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```
!  
dial-peer voice 100 voip  
  description Outbound to CUCM  
  translation-profile outgoing CUCM  
  session protocol sipv2  
  session target ipv4:192.168.100.200  
  voice-class sip transport switch udp tcp  
  voice-class sip conn-reuse  
  voice-class sip rel1xx disable  
  voice-class sip session refresh  
  voice-class sip midcall-signaling block  
  voice-class sip early-media update block  
  dtmf-relay rtp-nte  
  codec g711ulaw  
  no vad  
!
```

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Cisco 300-815 is an exam that focuses on the implementation of advanced call control and mobility services in Cisco's network infrastructure. 300-815 exam is designed to test the candidate's knowledge and skills in implementing and maintaining advanced call control and mobility services, including Unified Communications Manager, Cisco Unity Connection, and Cisco Unified IM and Presence.

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## What Career Prospects & Salaries Are Awaiting You?

Once you get CCNP Collaboration certified, you may opt for numerous jobs in large organizations and companies. As a rule, these positions can give the opportunity to earn higher salaries and develop professionally. Thus, you can apply for the following roles as rendered below:

- Voice and Network Specialist: \$74,868;
- Unified Communications Engineer: \$77,567.
- Network Engineer: \$74,768;

The above-mentioned average yearly remuneration is given as seen on the PayScale.com website. Plus, the same site claims that the annual salary for the CCNP certificate is almost \$97k.

## Cisco Implementing Cisco Advanced Call Control and Mobility Services Sample Questions (Q192-Q197):

### NEW QUESTION # 192

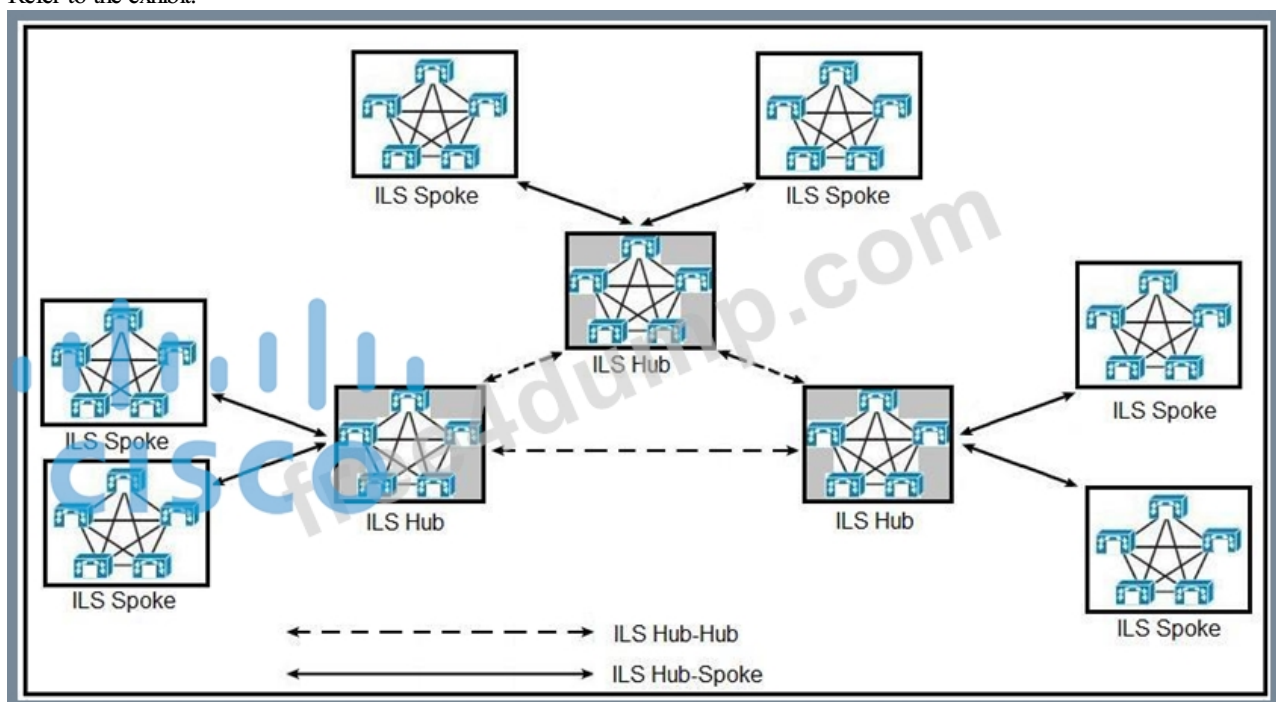
Which two extended capabilities must be configured on dial peers for fast start-to-early media scenarios (H.323 to SIP interworking)? (Choose two.)

- A. DTMF
- B. AUDIO
- C. FAX
- D. VIDEO
- E. BFCP

Answer: A,E

### NEW QUESTION # 193

Refer to the exhibit.



How many maximum hops can an ILS update traverse?

- A. 0
- B. 1
- C. 2
- D. 3

Answer: A

### NEW QUESTION # 194

Which section under the Real-Time Monitoring Tool allows for reviewing the call flow and signaling for a SIP call in real time?

- A. Voice/Video > Session Trace Log View > Real Time Data
- B. Analysis Manager > Inventory > Trace File Repositories
- C. Voice/Video > Session Trace Log View > Open From Local Disk
- D. System > Tools > Trace and Log Central

Answer: A

Explanation:

Reference:

<https://www.cisco.com/c/en/us/support/docs/unified-communications/unified-communications-manager-callmanager/213583-procedure-to-analyse-call-flow-of-sip-ca.html>

#### NEW QUESTION # 195

A company wants to use a Cisco phone system without the need for a local generic server. It is maximum 30 phones using a vendor-independent, industry-standard call protocol. What is the first command that must be entered to add this type of phone when registering a Cisco phone to this device?

- A. voice register sip ephone 3
- B. voice register ephone 3
- C. ephone 3
- D. voice register pool 3

Answer: D

#### NEW QUESTION # 196

Cisco SIP IP telephony is implemented on two floors of your company. Afterward, users report intermittent voice issues in calls established between floors. All calls are established, and sometimes they work well, but sometimes there is one-way audio or no audio. You determine that there is a firewall between the floors, and the administrator reports that it is allowing SIP signaling and UDP ports from 20000 to 22000 bidirectionally.

What are two possible solutions? (Choose two.)

- A. Ask the firewall administrator to change the range of UDP ports to 16384-32767.
- B. Go to the SIP profile assigned to these IP phones in Cisco Unified CM and change the range of media ports to 20000-22000.
- C. Go to the SIP profile assigned to these IP phones in Cisco Unified CM and change the range of media ports to 16384-32767
- D. Ask the firewall administrator to change the ports to TCP.
- E. Go to System Parameters in Cisco Unified Communications Manager and change the range of media ports to 20000-22000.

Answer: A,C

Explanation:

Section: Signaling and Media Protocols

Explanation/Reference: [https://www.cisco.com/c/en/us/td/docs/voice\\_ip\\_comm/cucm/port/9\\_1\\_1/](https://www.cisco.com/c/en/us/td/docs/voice_ip_comm/cucm/port/9_1_1/CUCM_BK_T2CA6EDE_00_tcp-port-usage-guide-91/CUCM_BK_T2CA6EDE_00_tcp-port-usage-guide-91_chapter_01.html)

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#### NEW QUESTION # 197

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