

CCNA Voice Bible (640-460) – Other Questions

Question 1 A SIP Trunk is a logical connection between an IP PBX and a Service Provider's application servers that allows voice over IP traffic to be exchanged between the two. When a call is placed from an internal phone to an external number, the PBX sends the necessary information to the SIP Trunk provider who establishes the call to the dialed number and acts as an intermediary for the call. All signaling and voice traffic between the PBX and the provider is exchanged using SIP and RTP protocol packets over the IP network. Which two statements about SIP trunk are true? A. A SIP trunk configuration is mandatory for a UC500 device. B. A SIP trunk is needed for internet access C. A SIP trunk is needed only for voice if you are planning on using VoIP through a service provider. D. A SIP trunk is not supported in a keyswitch configuration. Answer: C D Question 2 What is the default VTP mode on Cisco switches? A. Master B. Client C. Backup D. Server Answer: D Question 3 What are the three VTP modes on Cisco switches? A. Master B. Client C. Server D. Transparent Answer: B C D Question 4 What protocol needs to be enabled on an ATA if a fax machine is connected to the ATA? A. MGCP B. SCCP C. H323 D. SIP Answer: C Question 5 When would you require a voice gateway? (Choose two) A. When you conned a branch location to IP WAN B. When you connect a branch location using VoIP to the PSTN C. When you connect a Cisco Unified Communications Manager to a LAN D. When you connect a Cisco Unified Communications network to a PBX Answer: B D