[Lead2pass New Free Lead2pass Cisco 400-051 PDF Dumps With The Latest Update Exam Questions (321-340)

2017 November Cisco Official New Released 400-051 Dumps in Lead2pass.com! 100% Free Download! 100% Pass Guaranteed!
Lead2pass dumps for 400-051 exam are written to the highest standards of technical accuracy, provided by our certified subject matter experts and published authors for development. We guarantee the best quality and accuracy of our products. We hope you pass the exams successfully with our practice test. With our Cisco 400-051 dumps, you will pass your exam easily at the first attempt. You can also enjoy 365 days free update for your product. Following questions and answers are all new published by Cisco Official Exam Center: https://www.lead2pass.com/400-051.html QUESTION 321A company is decommissioning a site where a Cisco Unity Connection cluster resides. This cluster is part of a larger network of Unity Connection servers linked using HTTPS networking. Which three steps remove the site from the network? (Choose three.) A.   Determine if the Unity Connection cluster to be decommissioned sits between the hub and another Unity Connection site in the hub-and-spoke topology.B.   Remove the Unity Connection primary server from the HTTPS network on each node in the cluster.C.   Remove all servers in the Unity Connection cluster from the other clusters in the HTTPS network.D.   Update any downstream Unity Connection locations so that they link with a Unity Connection that will continue to have access to the hub location.E.   Remove the existing link to the remaining Unity Connection locations subtree and add new links to locations that will remain connected to the hub.F.   Update any remote call handlers and interview handlers that targeted the users on the location as well as any location downstream from the commissioned site to be removed.Answer: AEF QUESTION 322Refer to the exhibit. A Cisco collaboration engineer is writing a report to summarize the call distribution characteristics in a Cisco Unified Contact Center Express queue. Which three characteristics can be reported about the call distribution? (Choose three.)   A.   This queue will not work because no prompt has been selected.B.   Calls to this queue can be distributed in a round-robin manner between agents.C.   Agents that are answering calls for this queue can answer calls to other queues if available.D.   Agents in this queue are expected to finish (wrap-up) a call within 60 seconds.E.   Calls to this queue are handled in the order they were received unless prioritized by the script.F.   Changing the queue name from SupportQueue to Support01 requires updates to the script.G.   Agents logged in to this queue automatically receive calls without the need to do anything else (automatic work). Answer: CEF QUESTION 323A Cisco Unified Contact Center Express manager wants to add database integration to the self- service interactive voice response application. Which four types of licensing and database servers support this requirement? (Choose four.) A.   The server must have enhanced licensing.B.   The server must have premium licensing.C.   A server running Sybase Adaptive Server is required.D.   A server running Oracle is required.E.   A server running Postgress SQL is required.F.   A server running SAP SQL server is required.G.   A server running Microsoft SQL server is required.H.   The server must have standard licensing. Answer: BCDG QUESTION 324Multiple Jabber for Windows users are having problems logging into the voicemail server. The Cisco Unity Connection administrator has reset the password and emailed them the new credentials, as well as the instructions about how to reset them in Jabber. The users cannot see the Phone Accounts tab under Jabber settings to complete the instructions. Which two steps resolve this issue? (Choose two.) A.   In the Cisco Unified CM Jabber Service Profile, change the Credentials source for voicemail service to "not set".B.   In Cisco Unified CM, create a MailStore service and assign it to the Jabber Service Profile as Primary.C.   In the IM&P server CCMCIP Profile, uncheck the "Make this the default CCMCIP Profile for the system".D.   In the IM&P server Enterprise Parameters Configuration, enable the Phone Personalization parameter.E.   In the Cisco Unified CM Jabber Service Profile, uncheck "Make this the default service profile for the system". Answer: AB QUESTION 325A customer has a single Active Directory domain with users in various email domains. Each user is associated to only one email domain. The customer wants their users to federate to external organizations using their email addresses. What two methods are used to set up the integration between Active Directory, Cisco Unified CM, and IM&P? (Choose two.) A.   CUCM LDAP Attribute for User ID set to sAMAccountName, CUCM LDAP Directory URI set to mail, IM Address Scheme set to Directory URIB.   CUCM LDAP Attribute for User ID set to mail, IM Address Scheme set to User IDC.   CUCM LDAP Attribute for User ID set to sAMAccountName, CUCM LDAP Directory URI set to msRTCSIP-primaryuseraddress, IM Address Scheme set to Directory URID.   CUCM LDAP Attribute for User ID set to mail, CUCM LDAP Directory URI set to mail, IM Address Scheme set to Directory URIE.   CUCM LDAP Attribute for User ID set to mail, IM Address Scheme set to mail Answer: AD QUESTION 326A collaboration engineer is designing an 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 databaseB.   MySQL databaseC.   Microsoft SQL databaseD.   Postgree SQL databaseE.   Informix SQL database Answer: AD QUESTION 327Refer to the exhibit. A PSTN caller initiates an inbound call. Which two dial peers can be selected as inbound dial peers? (Choose two.)   A.   dial-peer voice 100 potsanswer-address [2-9]..[2-9]...$voice-port 0/0/0:23B.   dial-peer voice 200

potsdestination-pattern [2-9]..[2-9]..[2-9]...$voice-port 0/1/0:15C.   dial-peer voice 300 potsincoming called-number 704[2-9]...$voice-port 0/1/0:15D.   dial-peer voice 400 potsanswer-address 672[2-9]...$voice-port 0/0/0:15E.   dial-peer voice 500 potsincoming called-number 6..[2345689]...$voice-port 0/1/0:15 Answer: BD QUESTION 328Refer to the exhibit. A Cisco Unified CME administrator is configuring SNR for a line and has these requirements: -The remote phone should receive the call after the local phones ring for 10 seconds.-The ANI displayed on the remote phones should be the local extension number. Which two configuration commands complete these requirements? (Choose two.)   A.   snr 92875421 delay 15 timeout 10B.   snr 92875421 delay 10 timeout 20C.   snr calling-number localD.   snr calling-number remoteE.   snr answer-too-soon 10 Answer: BC QUESTION 329Refer to the exhibit. A call is received on a Cisco Unified Border Element gateway. The debug captures this SIP INVITE message. Which configurations result in 915556781234 as the final called number?   A.   voice class uri 1 siphost ipv4:10.41.11.10!voice class uri 2 siphost ipv4:10.50.40.30!dial-peer voice 1 voipincoming uri via 1translation-profile incoming 1!dial-peer voice 2 voipincoming uri via 2translation-profile incoming 2!dial-peer voice 3 voipincoming called-number +1Ttranslation-profile incoming 3B.   voice class uri 1 siphost dns:customer.comvoice class uri 2 siphost ipv4:10.50.40.30dial-peer voice 1 voipincoming uri via 1translation-profile incoming 1dial-peer voice 2 voipincoming uri via 2translation-profile incoming 2dial-peer voice 3 voipincoming called-number +1Ttranslation-profile incoming 3C.   voice class uri 1 siphost ipv4:10.41.11.10voice class uri 2 siphost ipv4:10.50.40.30dial-peer voice 1 voipincoming called-number +1[2-9]..[2-9]......translation-profile incoming 3dial-peer voice 2 voipincoming uri via 2translation-profile incoming 2dial-peer voice 3 voipincoming called-number [2-9]..[2-9]......translation-profile incoming 1D.   voice class uri 1 siphost dns:customer.comvoice class uri 2 siphost ipv4:10.50.40.30dial-peer voice 1 voipincoming uri via 1translation-profile incoming 1dial-peer voice 2 voipincoming uri via 2translation-profile incoming 1dial-peer voice 3 voipincoming called-number +1Ttranslation-profile incoming 3E.   voice class uri 1 siphost ipv4:10.41.11.10voice class uri 2 siphost ipv4:10.50.40.30voice class uri 3 siphost dns:customer.comdial-peer voice 1 voipincoming uri via 1translation-profile incoming 2dial-peer voice 2 voipincoming uri via 2translation-profile incoming 1dial-peer voice 3 voipincoming uri via 3incoming called-number +1Ttranslation-profile incoming 1F.   voice class uri 1 siphost ipv4:10.41.11.10voice class uri 2 siphost ipv4:10.50.40.30voice class uri 3 siphost dns:customer.comdial-peer voice 1 voipincoming uri via 1translation-profile outgoing 3dial-peer voice 2 voipincoming uri via 2translation-profile incoming 2dial-peer voice 3 voipincoming called-number +1Ttranslation-profile incoming 2 Answer: B QUESTION 330An engineer received this requirement from a service provider: Diversion header should match the network DID "123456@company.com" for Call Forward and transfer scenarios back to PSTN. Which SIP profile configuration satisfies this request? A.   voice class sip-profiles 200request INVITE sip-header Diversion modify "sip:(.*>)" "123456@company.com>" request REINVITE sip-header Diversion modify "sip:(.*>)" "123456@company.com>"B.   voice class sip-profiles 200request INVITE sdp-header Diversion modify "sip:(.*>)" "123456@company.com>" request REINVITE sdp-header Diversion modify "sip:(.*>)" "123456@company.com>"C.   voice class sip-profiles 200response 200 sdp-header Diversion modify "sip:(.*>)" "123456@company.com>"D.   voice class sip-profiles 200response 200 sip-header Diversion modify "sip:(.*>)" "123456@company.com>" Answer: A QUESTION 331Refer to the exhibit. Which three options are potential reasons the FXO port is not receiving Caller ID? (Choose three.)   A.   "Enable Caller ID" was not configured on the Cisco Unified Communications Manager configuration.B.   The FXO port was configured to "loop-start" instead of "ground-start".C.   The Timing Guard-out parameter is incorrectly set to 1500 ms.D.   "Connection polar opx immediate" was used and does support caller ID.E.   Gateway with IOS 12.4(24T) was used and does not support this feature.F.   The NTFY message contains a Hung Up parameter. Answer: ACE QUESTION 332Refer to the exhibit. Which three Ethernet Setup Administrator Settings are manually configurable locally on the Cisco 9971 IP phone? (Choose three)   A.   Operational VLAN IdB.   Admin VLAN IdC.   PC VLAND.   SW Port SetupE. PC Port Setup Answer: BDE QUESTION 333Refer to the exhibit. What caused this message on a Cisco 9971 IP phone, connected to a Cisco 3750X PoE switch, when a mobile phone is plugged into the IP Phone's back USB port?   A.   The back USB port only supports Cisco USB devices such as a Cisco Unified Video cameraB.   USB classes for this USB port are not properly configuredC.   The USB port is not enabled by the administratorD.   The mobile phone is requesting more power than the USB port could provideE.   USB devices are not supported when the IP phone is powered by a Cisco PoE switch Answer: D QUESTION 334Under which three conditions will a Cisco 9971 IP Phone request the "xmldefault.cnf.xml" file from a TFTP server in a Cisco Unified CM cluster? (Choose three) A.   The phone is registered to the CUCM cluster but need to update its firmwareB.   The phone is attempting to register to the CUCM cluster for the first timeC.   Auto-registration is disabled on CUCM clusterD.   The phone has not yet been defined in the CUCM databaseE.   The phone is attempting to change from SIP firmware to SCCP firmwareF. Auto-registration is enabled on CUCM cluster Answer: BDF QUESTION 335Which two options are Power Save configuration parameters for Cisco 9971 IP Phones? (Choose two.) A.   Phone On TimeB.   Phone Off Idle TimeoutC.   Day Display Not

ActiveD.   Enable Audio AlertE.   Enable Power Save PlusF.   Display on Duration Answer: CF QUESTION 336Which Cisco Unified CM Application user is created by default and used by Cisco Unified CM Extension Mobility? A.   CCMAdministratorB.   EMSysUserC.   TabSyncSysUserD.   CCMSysUserE.   CTIGWUser Answer: DExplanation: http://www.cisco.com/c/en/us/td/docs/voice_ip_comm/cucm/srnd/7x/uc7_0/directry.html   QUESTION 337Tom Lee is an active user in a Cisco Unified CM deployment with fully functional LDAP synchronization and authentication to an Active Directory. Daily resynchronization is set at 11:00 pm. At 8:00 am on March 1st 2014, this user was deleted from the AD.What will Tom Lee experience when he attempts to log in Extension Mobility at an IP phone and then access his Unified CM User Options page on his PC, at 9:00 am on March 1st 2014? A.   Extension Mobility will not work, but the User Options page will work.B.   Extension Mobility and the User Options page will not work.C.   Extension Mobility will work, but the User Options page will not work.D.   Extension Mobility and the User Options page will work.E.   The information provided is insufficient to answer this question. Answer: C QUESTION 338Tom Lee is an active user in a Cisco Unified CM deployment with fully functional LDAP synchronization and authentication to an Active Directory. Daily resynchronization is set at 11:00 pm. At 8:00 am on March 1st 2014, this user was deleted from the AD. What is the status of this user in the Unified CM database at 1:00 am on March 2nd 2014? A.   activeB.   inactiveC.   delete pendingD.   awaiting authorizationE.   permanently deleted Answer: B QUESTION 339Which option is a mandatory LDAP attribute for a user to be synchronized to Cisco Unified Communications Manager? A.   uidB.   telephoneNumberC.   employeeNumberD.   snE.   mail Answer: D QUESTION 340Which three softkeys can be offered on a Cisco IP Phone 7965, running SCCP firmware, when it is in On Hold state? (Choose three.) A.   SelectB.   ConfrnC.   NewCallD.   EndCallE.   iDivertF.   ParkG.   Hold Answer: ACE We offer standard exam questions of Cisco 400-051 dumps. The standard exams are important if you have never taken a real exam. The accuracy of the Q&As are fully guaranteed and the number is enough to impact you passing the exam. 400-051 new questions on Google Drive: https://drive.google.com/open?id=0B3Syig5i8gpDcVpjV1ZNcjVzaW8   2017 Cisco 400-051 exam dumps (All 542 Q&As) from Lead2pass: http://www.lead2pass.com/400-051.html [100% Exam Pass Guaranteed]