## SBST (Survivable Remote Site Telephony) 10 CLICM Call Flows

SRST (Survivable Remote Site	Telephony)				
What topology can use SRST	Centralized Remote Branch Call Flow—signalling always goes from remote branch(es) to a central CUCM server, while the audio only hits the WAN if needed				
What triggers SRST	WAN (and contact to the centalized CUCM server) is lost.				
How does SRST handle calls from the remote site	Router takes over, numbers to work	nding the calls over PSTN, modifying phone			
What does CUCM do with the branch office phones	Deregisters them because they're unreachable				
What is CFUR	Call Forward UnRegistered. An option in CUCM for each remote branch phone, giving a full PSTN number to be used to reach it if the phone ever becomes deregistered				
Centralized CUCM Deploymer	rt				
Maximum locations per cluster		2000			
Maximum H.323 or MGCP devices per cluster		2100			
Protocol where CUCM Limits simultaneous calls across WAN to QoS-allocated bandwidth		CAC (Call Admission Control)			
Protocol triggered only by above to redial excess calls to send them across PSTN		AAR (Automated Alternate Routing)			
Recap—WAN failure call handling		Hierarchical dial plan and CFUR send call to PSTN			
Recap—WAN full call handlin	ng	CAC triggers AAR, which redials the call with a full PSTN number			
Distributed Deployment					
Given site A with phone 1 and CUCMA and site B with phone 2 and CUCMB, what's the call audio flow for a connected call phone 1 to phone 2		<ul> <li>Phone 1</li> <li>(skip)</li> <li>(skip)</li> <li>Phone 2</li> </ul>			
Given the same topology, what's the call signaling flow to establish a call between phones 1 and 2		<ul> <li>Phone 1</li> <li>CUCMA</li> <li>CUCMB</li> <li>Phone 2</li> </ul>			
A router 10s feature that provi service to track available band clusters and kill or redial calls	des a <i>centralized</i> width between accordingly	Gatekeeper CAC			

## **Call Routing**

Possible Call Routing Destinations	<ul> <li>DN</li> <li>Translation Patterns</li> <li>Route Patterns</li> <li>Hunt Pilot</li> <li>Call Park Number</li> <li>Meet-Me Number</li> </ul>				
Possible Sources of Dialed Digits for Call Routing	<ul> <li>IP Phone</li> <li>Trunk</li> <li>Gateway</li> <li>Translation Pattern</li> <li>Voicemail Port (app ca</li> </ul>			Ills to notify msg recipient)	
Signaling protocol(s) between phones and CUCM	ocol(s) between phones • SCCP • SIP				
Two peer to peer gateway protocols	• H.323 • SIP				
Two gaveway control protocols	<ul><li>MGCP</li><li>SCCP</li></ul>				
Construct that groups similar devices like WAN or ISDN links	Route Groups				
A preference (cost?) -ordered list of above; can be used as a call destination	Route List				
Construct that matches dialed digits & gives it a destination	Route Pattern				
Call-Routing Behavior					
Are calls routed digit-by-digit or all-at-o	once	e All-at-once			
How choose between matching rules		Most specific (fewest wildcard possibilities)			
Hunt Group (set of extensions reached				a common number, e.g. helpdesk)	
List of participating extensions with ring down, circular, or broadcast) and when o	ging instr or if to giv	uctions (1 re up on t	top- his list	Hunt Group	
List of the above lists, always top-down				Hunt List	
Call routing entry invoked by one extens the above list of lists	sion num	ber that t	argets	Hunt Pilot	
Calling Restrictions					
A group of dialable things to be similarly restricted Partitions					

Note: Understand the two interactions under "Notes" on page 5 of the Chapter 10 reading notes.

Calling Search Space

List of the above groups, assigned to phones as allowed