

"AVVID"

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# " Overview " Announcing AVVID products " Design models " Design consideration " Case studies



#### Cisco AVVID—An End-to-End Architecture



Infrastructure

#### Cisco IP Fabric



Directory
Policy
Management





Scalable Open

**END TO END MANAGEMENT** 

www.cisco.com

Applications

Service

Call Mgr Servers

Telephony
Apps
Servers

Directory Servers

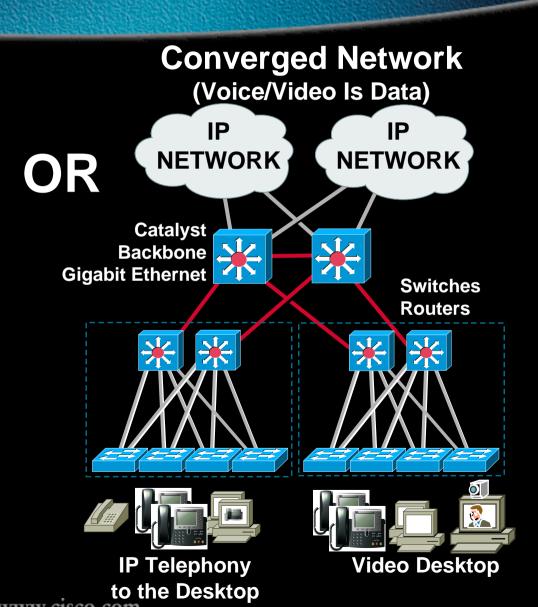
Message Servers

Content Servers

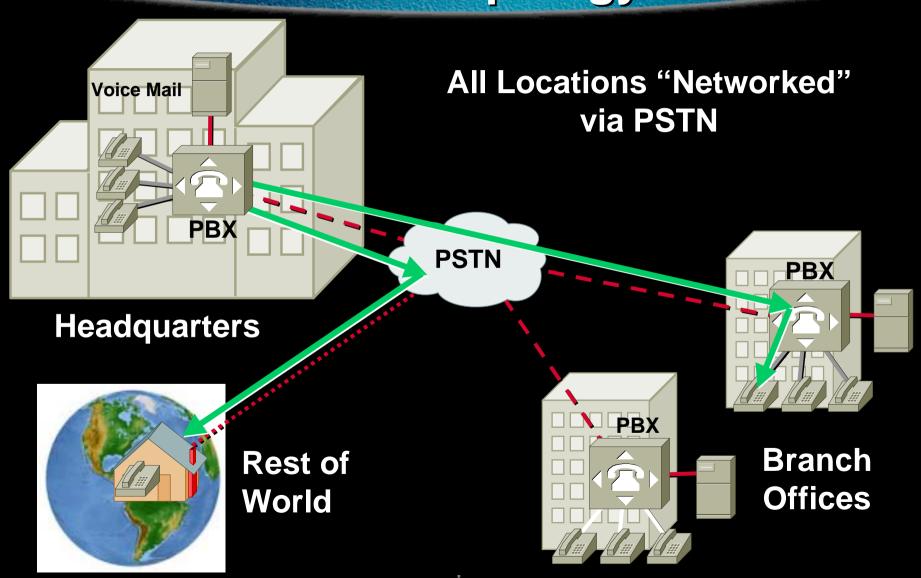
Wanagement

#### Infrastructure Choices

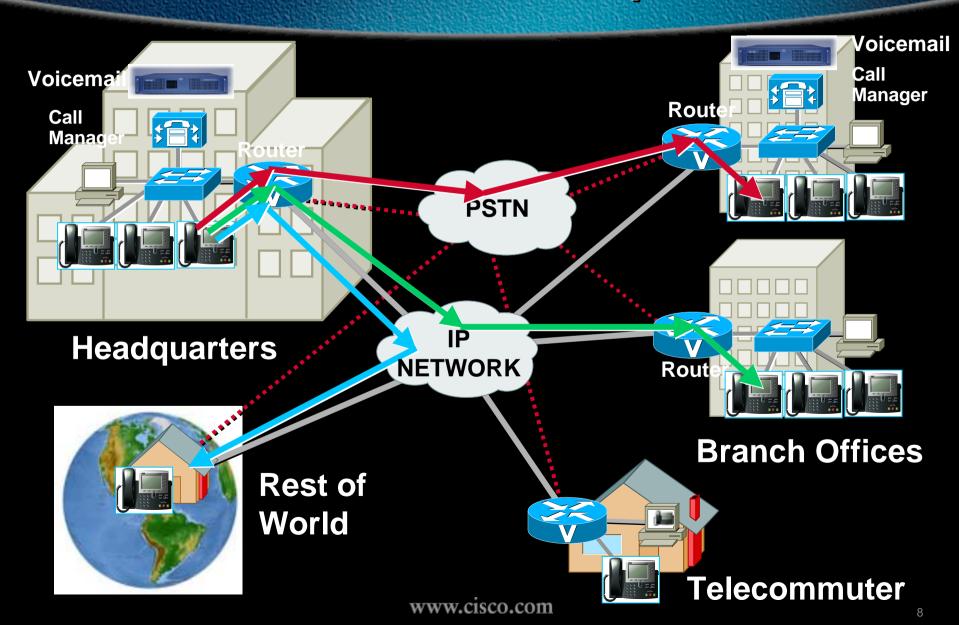
**Converged Network** (Data over Voice) **IP PSTN NETWORK PBX** 



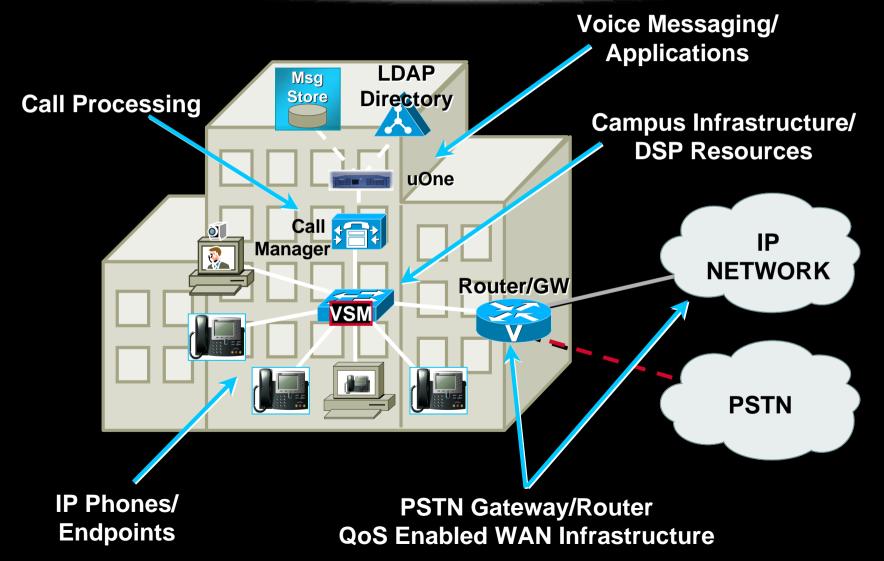
## Today's Enterprise Voice Topology



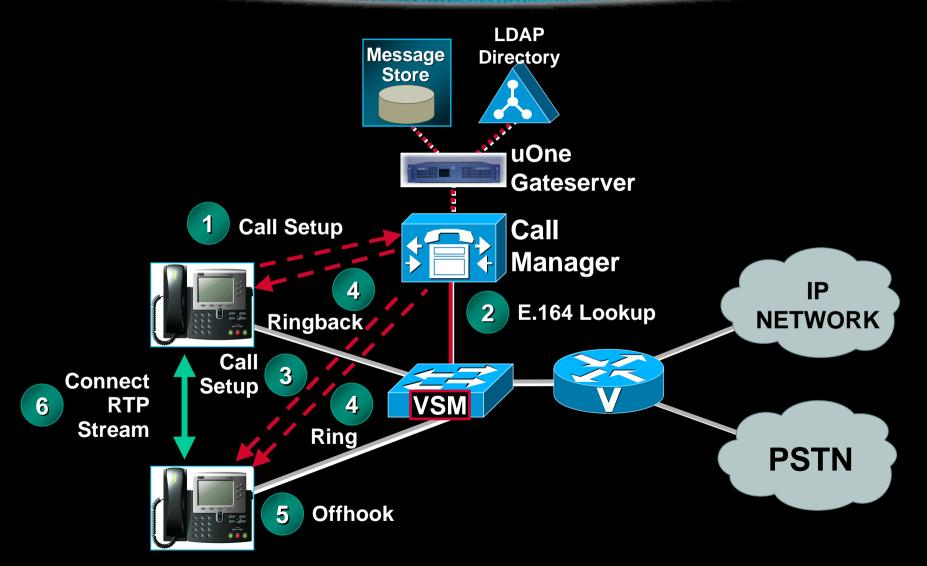
### **Tomorrow's Enterprise**



## Examining IP Telephony Components



### Call Processing





#### **IP PBX: MCS 7835**



- Call control and processing
- " Support 2500 IP phones
- " 10,000 IP phones per cluster
- " Dual 18.2 Gig hot plug drives
- Dual hot swappable power supply
- " 50,000 Busy Hours



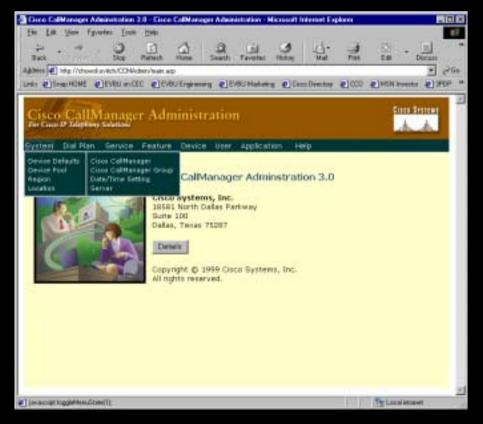
#### **IP PBX: MCS 7822**



- " Call control and processing
- " Single 9.1 Gig hard drive
- " Single power supply
- " Lower cost platform

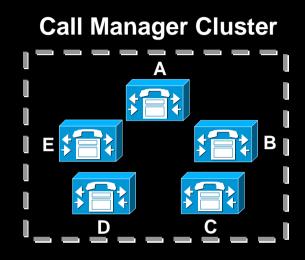


### Call Manager 3.0



- Standard HTML which minimizes training
- Pre-installed on Cisco platforms
- " Comprehensive telephony features
- Remote administration simplified
- Extensive on-line help
- " 50,000 Busy Hour Call Completion Setup per MCS server
- " 125,000 Busy Hour Call Completion Setup per cluster
- " Scalability, availability and redundancy

### Call Manager Cluster Scalability



#### **Call Manager Cluster Sizing**

- 1. Five Call Managers max in a cluster—cluster is confined to a campus
- 2. 2500 users max per Call Manager (even under failure conditions)
- 3. Maximum of 10,000 users in a cluster
- 4. Provision for Call Manager failure

#### **Call Manager Cluster IP Phone Provisioning**

CM's in Cluster	Max users per cluster	Max users with N+1 Redundancy
1	2500	0
2	5000	2500
3	7500	5000
4	10,000	7500
5	10,000	10,000

#### Cisco IP Phone

NEW

#### **Cisco** 7960



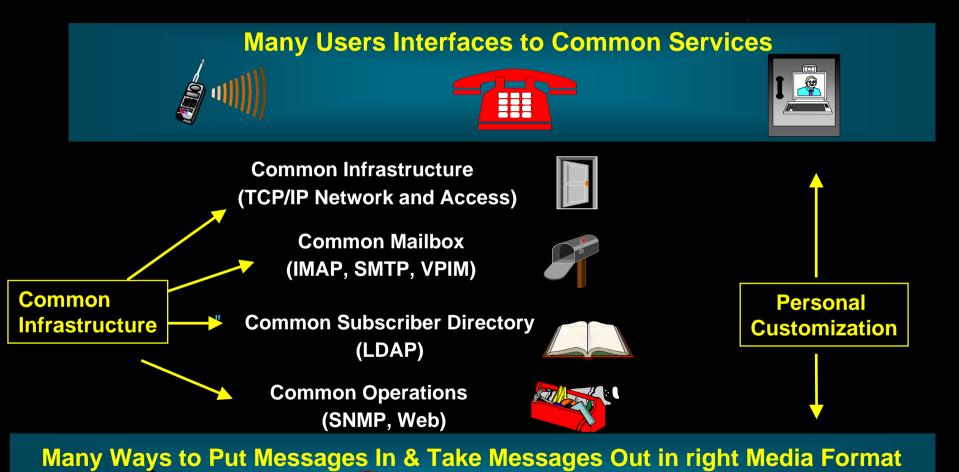
- Professional, manager
- Six lines—Mix directory numbers or features
- Display area: Calling information, feature access via "soft keys," full-duplex handsfree
- 10/100BaseT switch

#### **Cisco 7910**



- Common areas—hallway, break room, reception etc
- Single line
- Two models—Single 10Base T or 10/100Base T Ethernet
- FCS Q3 2000

## **Unified Messaging System**



## uOne Voice Mail uOne 4.1E Corporate Edition

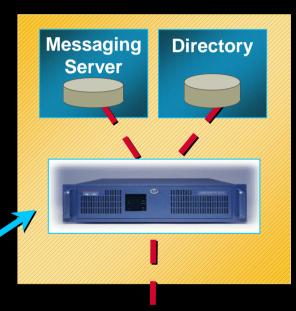
#### MCS-7835

- " uOne GateServer 4.1E
- " Embedded Message Server
- " Embedded Directory Server
- " 2 x 18GB Drives

#### uOne 4.1E—Corp

- **" 500 users Voice Mail Only**
- " 20 Simultaneous VM Ports
- " 30 Min of Voice Mail Storage per User

Voice-Mail MCS 7835

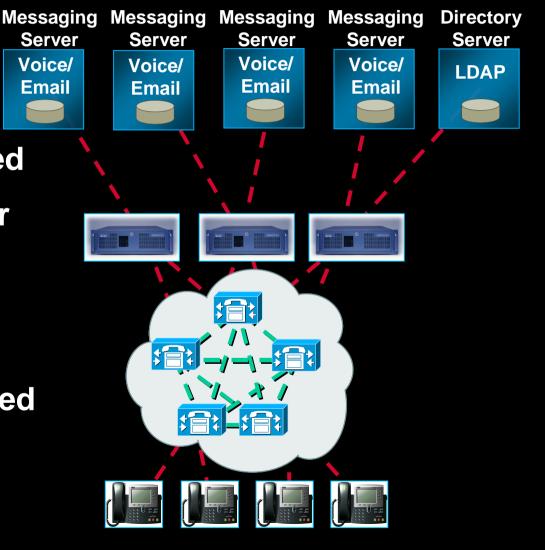




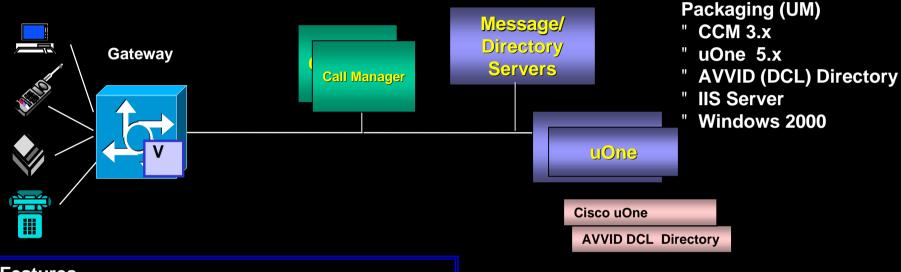
Call Manager 3.0

## uOne 5.0E—Corporate: 500 to 10,000 User Campus

- Voice mail networking supported
- " Call Manager failover supported
- " G.711 OR G.729
- Single, central directory or embedded AVVID directory
- " Scalable growth



#### uOne Unified Messaging



#### **Features**

- FAX mail, Email, Voice Mail unified
- **Scheduled Delivery, Cancelled Delivery**
- Media Conversion: Text-to-FAX, Text-to-Speech
- " VPIM/AMIS-A Message Interchange
- N+1 redundancy/recovery
- **XML Customizable TUI & Web Client**
- Internationalization
- **Single Number Reach Services**
- Notification via MWI, Paging, SMS or PDA

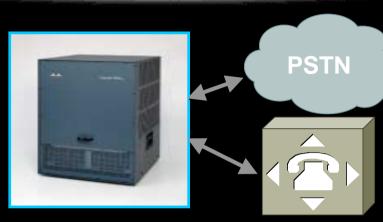
#### **Deployment Options**

- Separate CM, uOne, Message Servers
- Co-Resident uOne, CM and Message Server
- **Co-Resident uOne and Message Server** separate Call Manager

## Catalyst 6000 8-Port Voice Services Module (VSM)



Catalyst 6000

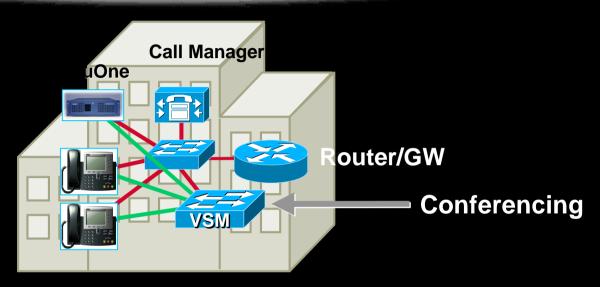


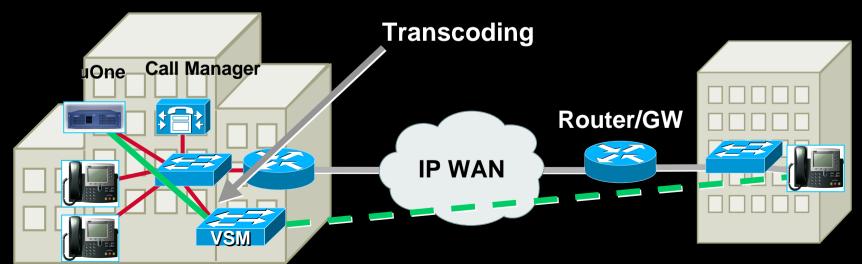


- " Catalyst 6000 8-port voice T1 and services module:
  - Provides eight physical or logical ports (configured on a port by port basis)
  - When configured as a physical port:
    - T1 or E1 gateway for connections to the PSTN or PBX (built in CSU)
    - **Supports ISDN PRI at FCS**
  - When configured as a logical port:
    - Provides voice services such as conferencing or transcoding

## Conferencing and Transcoding Resources

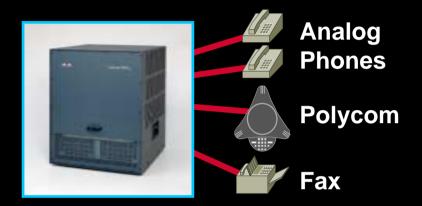






## Catalyst 6000 FXS ports

- Deployable in any Catalyst 6000/6500 chassis
- Provides analog connectivity for fax machines, speaker phones, analog telephone sets
- " 24 FXS ports (one RJ21 connector)
- " Uses one IP address
- Supports fax relay
- Supports hook-flash blind transfer

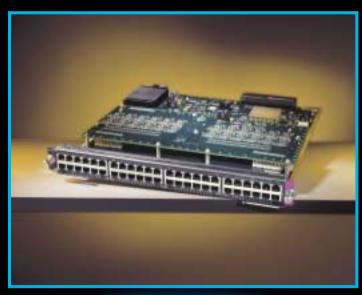




#### Catalyst 6000 Inline Power

- " 48-port 10/100 Ethernet switch line card, with optional Inline Power daughtercard
- Provides Inline Power to IP phones
- Intelligent auto-detection of IP phones
- " Manageable via SNMP





### Catalyst 3524-PWR-XL

#### **Enhanced Functionality:**

" QoS - ensure timely delivery of voice packets

Reclassification of 802.1p priority settings on a portby-port basis

High Availability - increase reliability of network

Phone power - integrated Inline Power for Cisco IP phones over the same pairs as Ethernet (pins 1,2,3,6)

New low-cost RPS using new power architecture

Integrated Management - manage network as one system

IP phone configuration Auxiliary VLAN



### Catalyst Inline Power Patch Panel

NEW

- For use in any switched Ethernet environment
- Provides Inline Power to any Cisco IP phone
- " Auto-discovery algorithm to provide power on demand to IP phones
- Increases availability of IP telephony network via centralized power distribution



#### VG200 Standalone Gateway

- " Second-generation standalone analog gateway
- Supports up to four FXS/FXO interfaces
- " Support E1/T1 in CQ3
- " Uses standard Voice Interface Cards (VIC) from router platforms
- " Voice gateway only; no routing capability





## Video Integration with the IP/VC Family

- " Cisco IP/VC 3510

  Multipoint control unit
- " Cisco IP/VC 3520, 3525 H.320 to H.323 gateways
- Cisco IP/VC 3530Video terminal adapter
- Cisco MultimediaConference ManagerH.323 Gatekeeper/proxy

Cisco 3500 Series

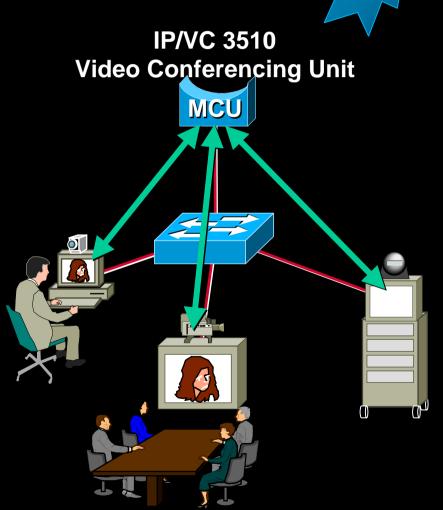




## Cisco IP/VC 3510 Multipoint Control Unit

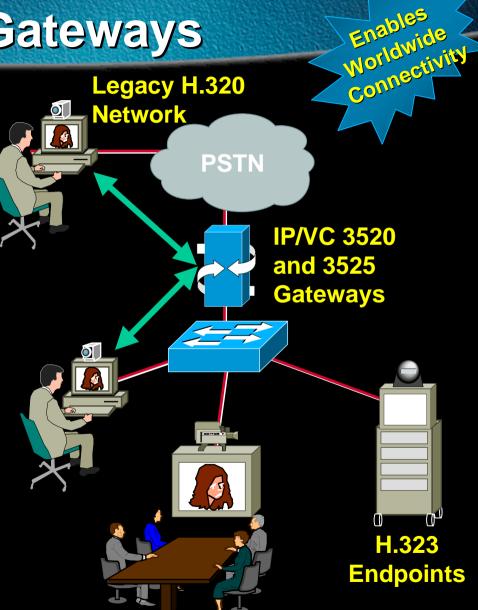
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- Enables multi-participant videoconferences
- Video, audio and T.120data bridging from 128kbps to 1.5 Mbps
- " Up to 15 sessions per unit
- " Scales easily by combining multiple units
- " Supports ad hoc call-in or planned call-out conferences



## Cisco IP/VC 3520 and 3525 Gateways

- Connects H.320 circuitswitched networks to H.323 IP networks
- Ties into existing H.320 conferencing networks
- Video, audio, and T.120 data calls up to 384 kbps
- Supports PRI, BRI and V.35 interfaces



## Cisco IP/VC 3530 Video Terminal Adapter

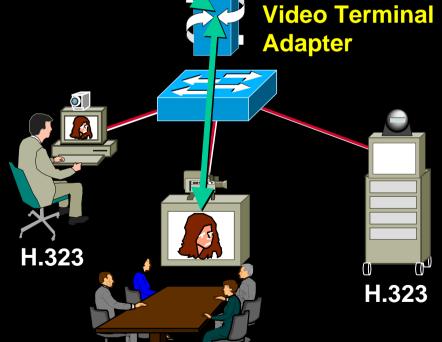
Network n investment

**IP/VC 3510** 

preserves

Connects an H.320 system to an H.323 IP network

- Video, audio, and T.120data calls up to 384 kbps
- Provides migration path to IP networks

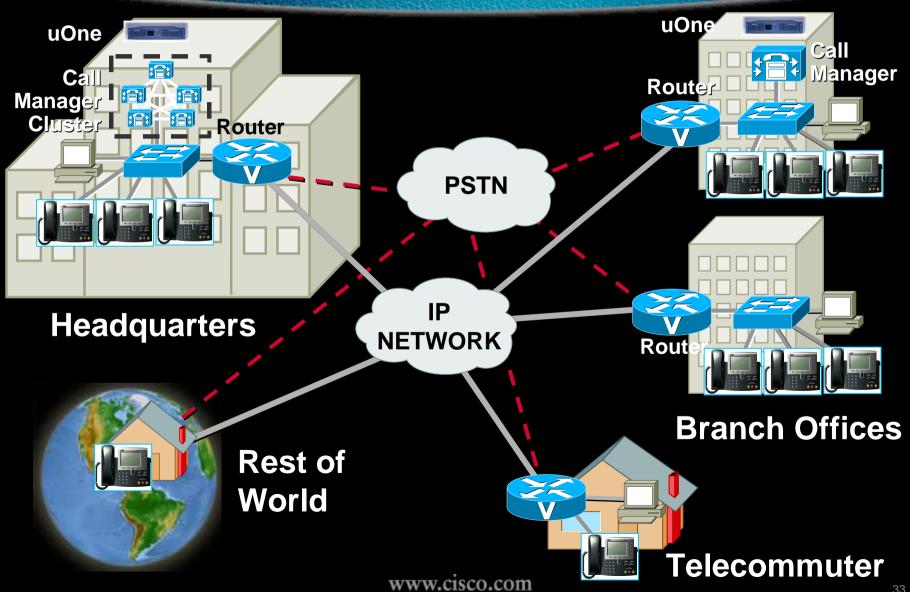


**PSTN** 

Legacy H.320

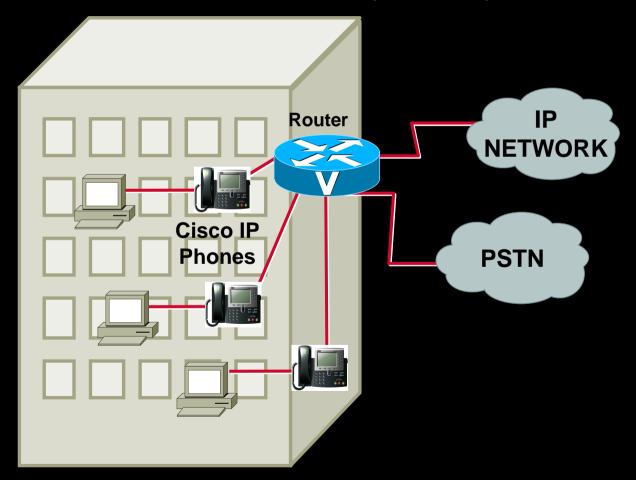


### What We Are Going to Build?

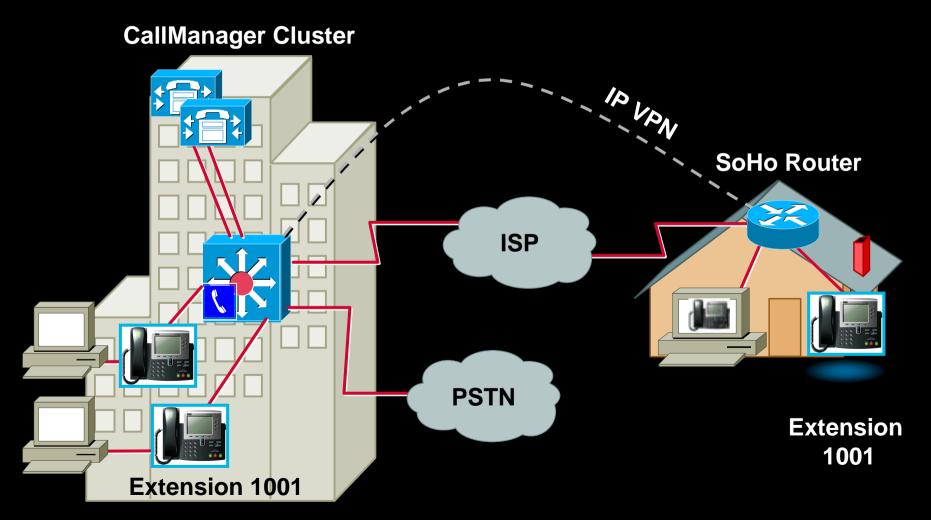


#### **Small/Branch Office**

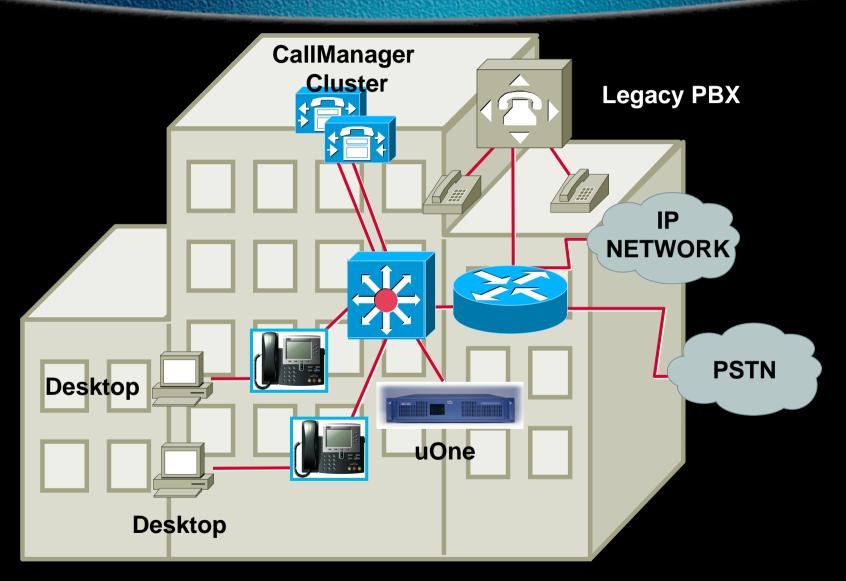
Single Infrastructure for Data, Voice, and Video



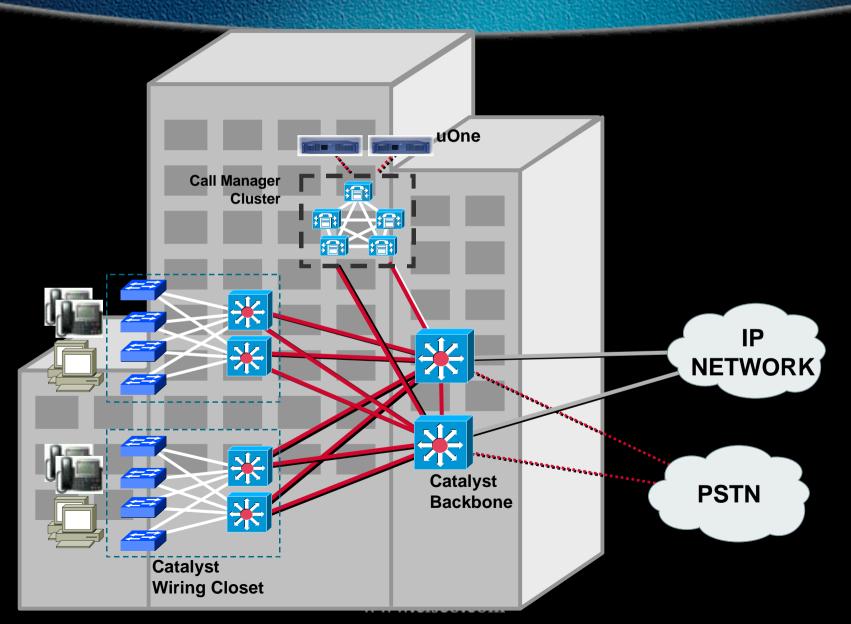
#### Telecommuter



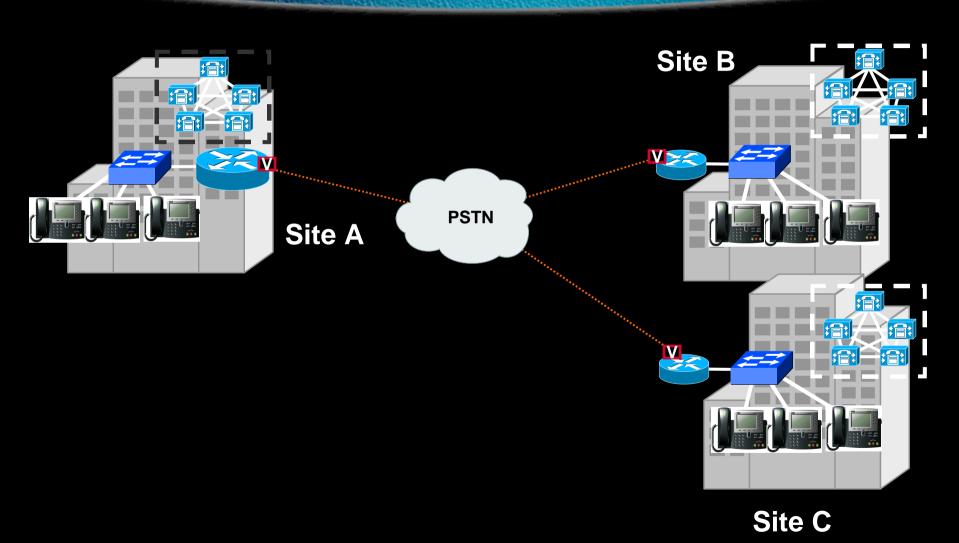
## **Converged Campus**



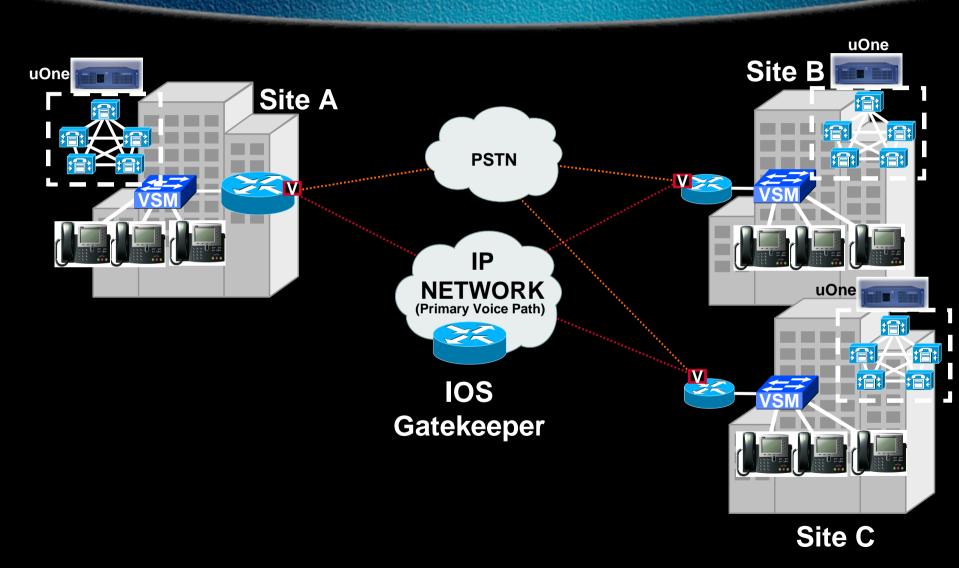
## **Headquarter Deployments**



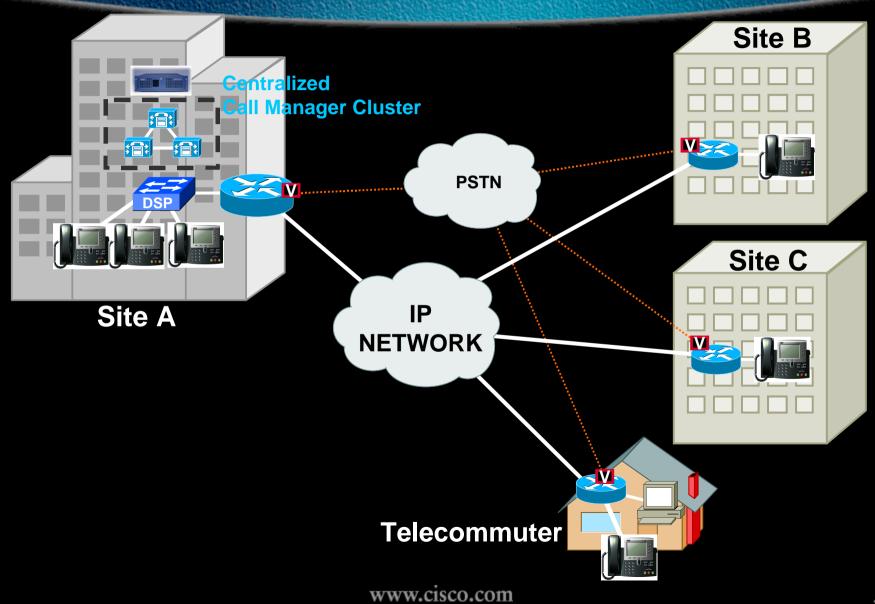
## **WAN - Isolated Deployments**



## WAN - Distributed Call Processing



## WAN - Centralized Call Processing





# Deployment Questions: Where Do I Start?

- " Redundancy/scalability/availability
- Power source
- " Cabling
- " IP addressing
- " Centralized vs. distributed call processing
- Interoperability
- " QoS (Quality of Services)
- Bandwidth provisioning

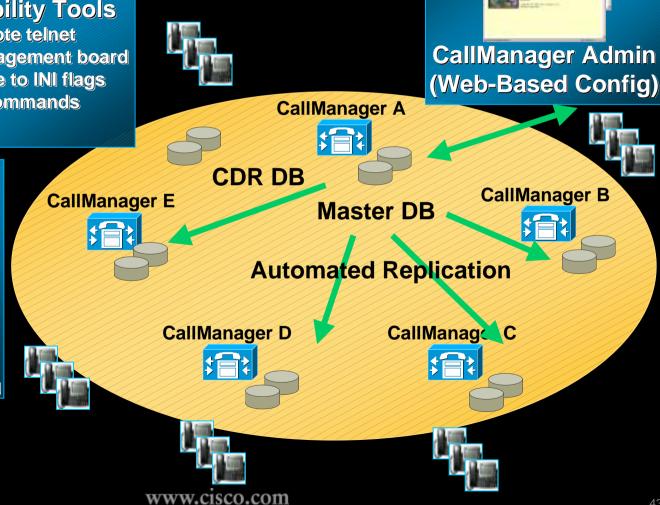
## Call Manager Cluster design

#### Integrated **Serviceability Tools**

- " Bridged remote telnet
- " Remote management board
- " Web interface to INI flags
- " Show tech commands
- " Carbon copy

#### Web-Based Management

- " Syslog
- " Device statistics
- " Signaling/control debug trace
- " CDR reporting
- " Event monitoring
- " Performance Monitoring



### Power

- Inline Power
  - Needs Powered Linecards for Catalyst Switches
  - ∠ Uses Pairs 2 & 3 (same as Ethernet) for delivering power
- External Power

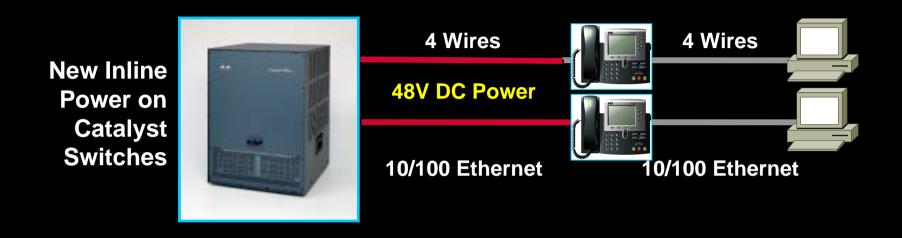
Wall Power

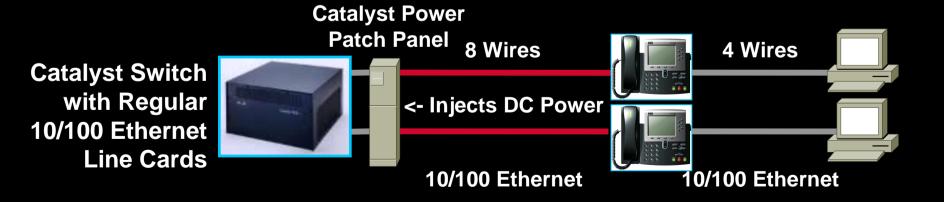
- Needs External Power Patch Panel
- Patch Panel delivers power over Pairs 1 & 4
  - AC SQUIED CONTROL OF C
- Needs DC Converter for connecting IP Phone to Wall Outlet

Inline power: Catalyst 6000, 6500, 4006, 3500-PWR

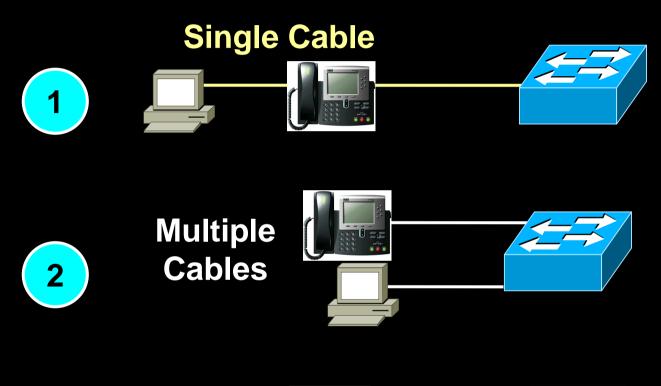


## Single Wire and Power Options





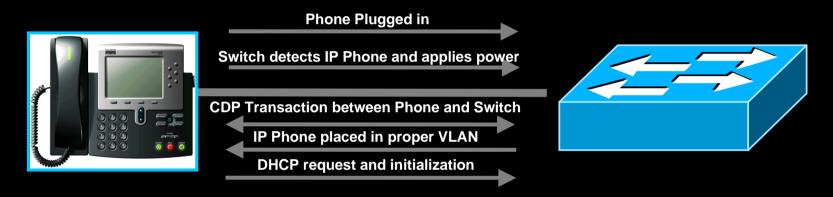
# Cabling





## Discovering Cisco IP phones

### **Endpoint/Infrastructure Integration**



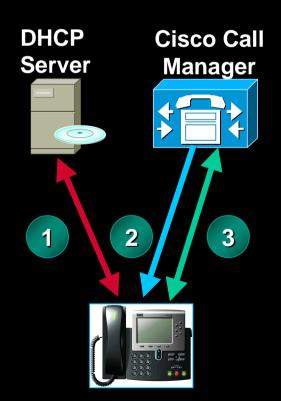
10/100 switch port
LDAP enabled
IP Prec = 5
802.1Q VLAN Support

24 Port Ethernet Line Card Uses Pins 1,2,3,6

**Catalyst 6K, 4K + 3500** 

## Flexible Mobility

- 1. Phones make DHCP request to get an IP address, gateway, boot server, etc.
- 2. Phones make TFTP boot file request to get CM IP addresses
- 3. Phones register with CM and get Display Templates and ready to receive/place calls



#### Ease of Moves, Adds, and Changes

- Add a new devicePlug it in out of the box
- Move a deviceUnplug and plug in new location
- Changes
  Simple web-based interface

## **IP address**



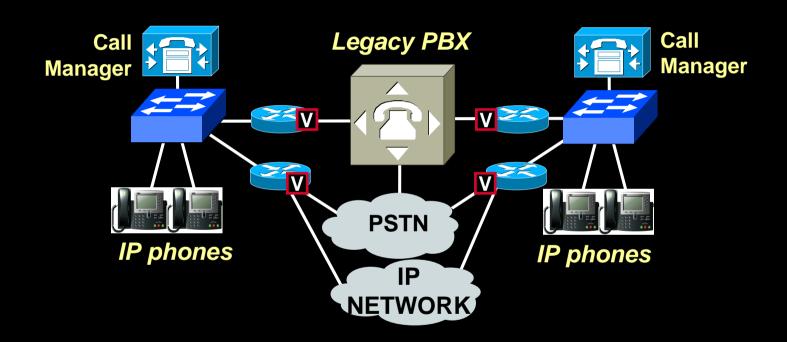






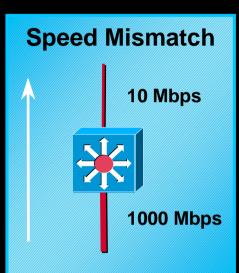


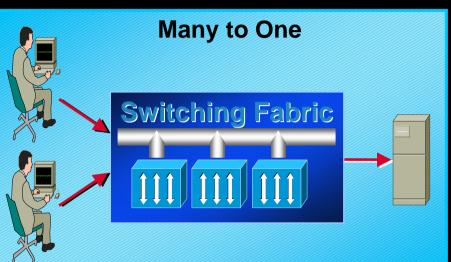
## Interoperability

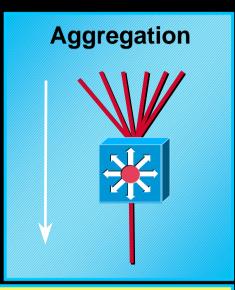


- " Signaling: E1/T1 CAS, CCS, QSIG, PRI, BRI
- "FXO, FXS, E&M
- " H323, MGCP
- " Standards based

## Campus QoS

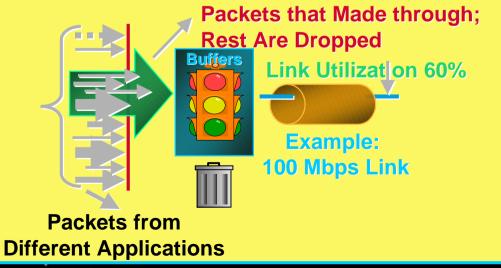




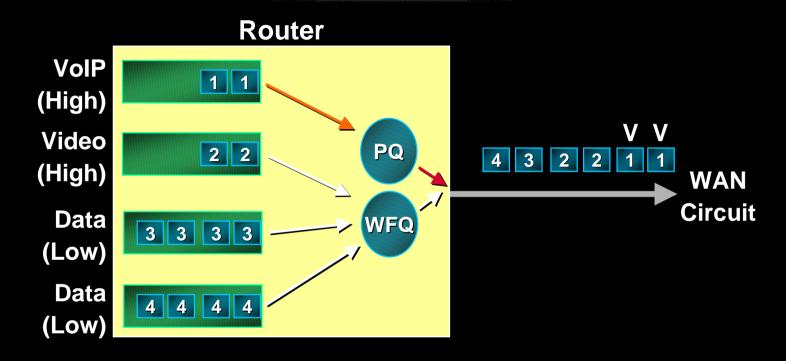


TAG voice packet
Voice VLAN
Separate queue for voice
Trusted boundary





# QOS in WAN Protecting Voice from Data



#### **Classification Mechanisms**

IP Precedence
DiffServ Code Point
RSVP

"Identifying" Voice as Important

#### **QoS Queuing Features**

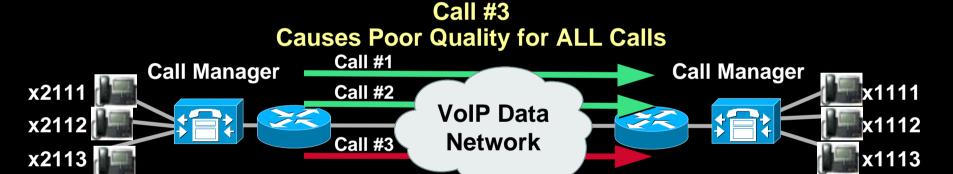
IP RTP Priority
Low Latency Queuing

"Giving Priority" to Voice

# QOS in WAN: Protecting voice from voice

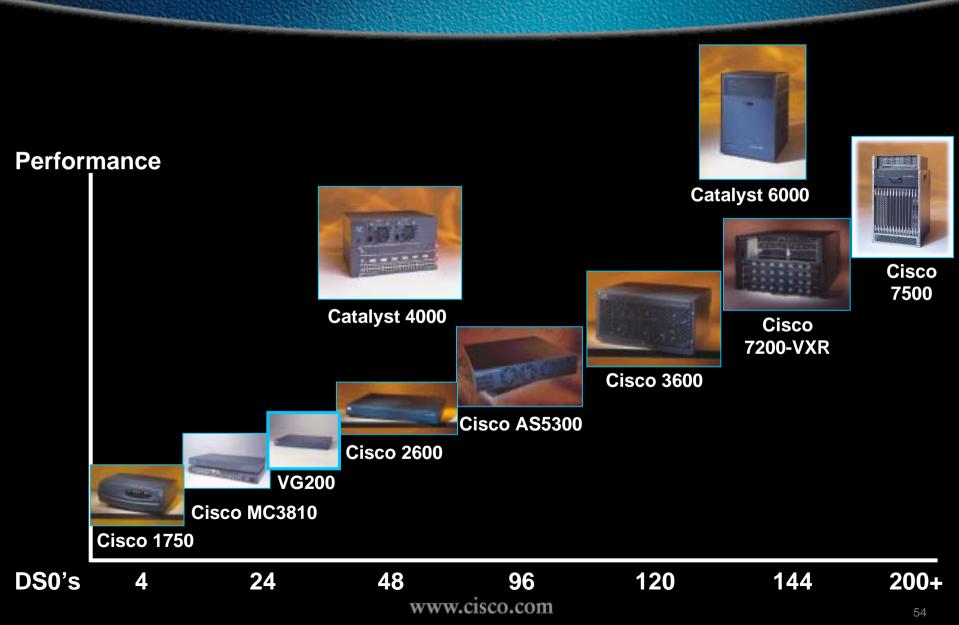
#### **Example:**

WAN Bandwidth Can Only Support 2 Calls What Happens when 3rd Call Attempted?



Use gatekeeper between Call Manager
Use Call Manager Location within Call Manager

## Cisco Integrated Voice Gateways



# Design considerations: WAN bandwidth requirement

How much bandwidth do I need for one voice call?

Link	IP Header	UDP	RTP	Voice		
Header		Header	Header	Payload		
X Bytes	20 Bytes	8 Bytes	12 Bytes	X Bytes		

" G711 64K

' **G72**9a 8K

"RTP/UDP/IP 16K

" CRTP 2K

" VAD uses less



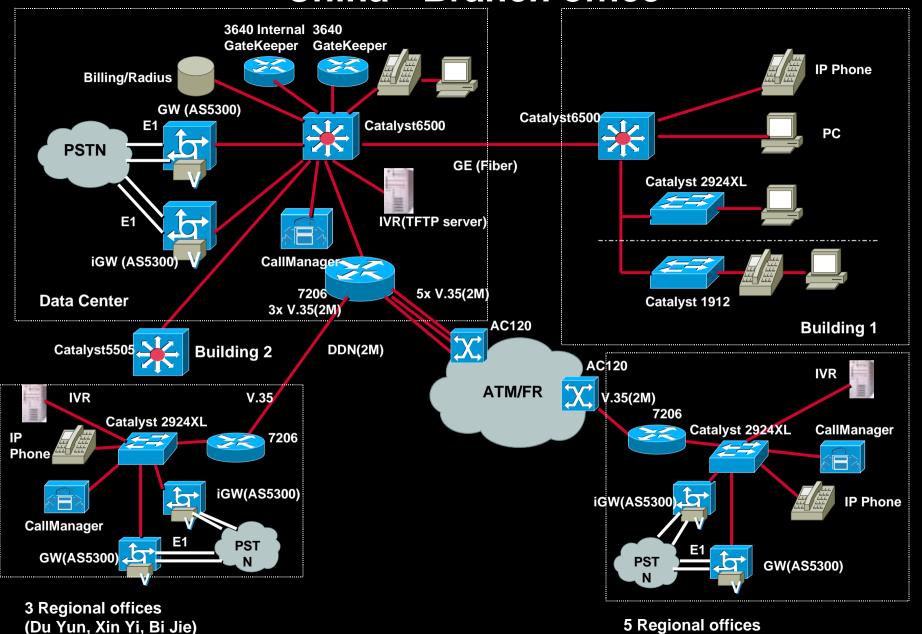
- " Use G711 in LAN
- " Use G729a in WAN
- " Consider transcoding from G711 to G729a
- " With CRTP and G729a use 10K





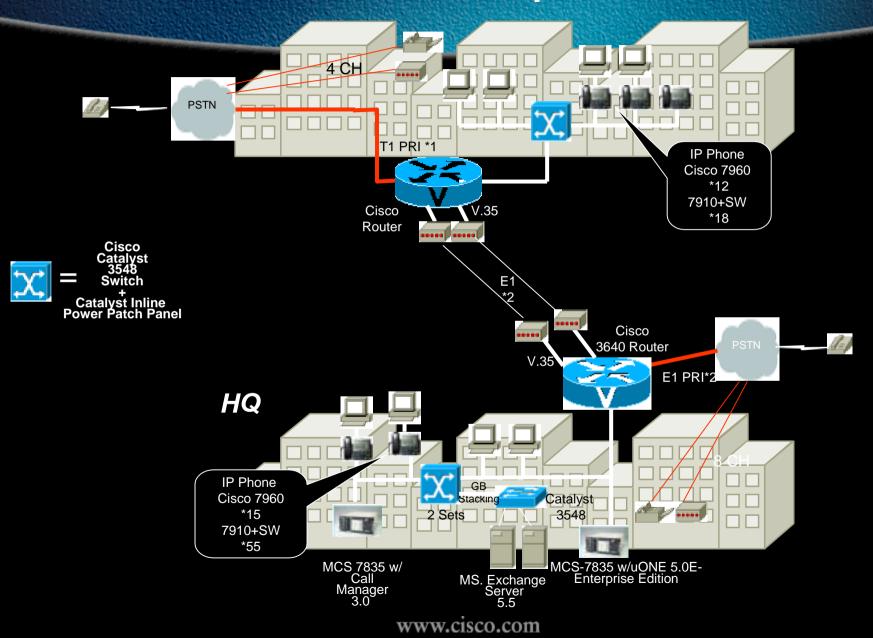


### **China - Branch office**



(Zun Yi, Kai Li, Liu Pan Shui, An Shun, Tong Ren)

### **Taiwan: Campus**

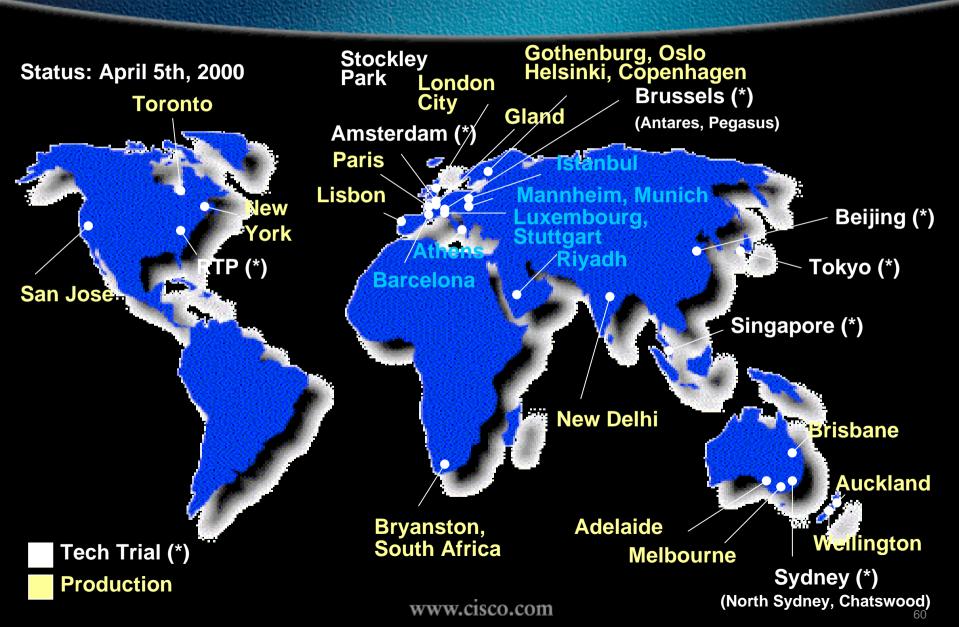


# Cisco Systems: Worldwide deployment

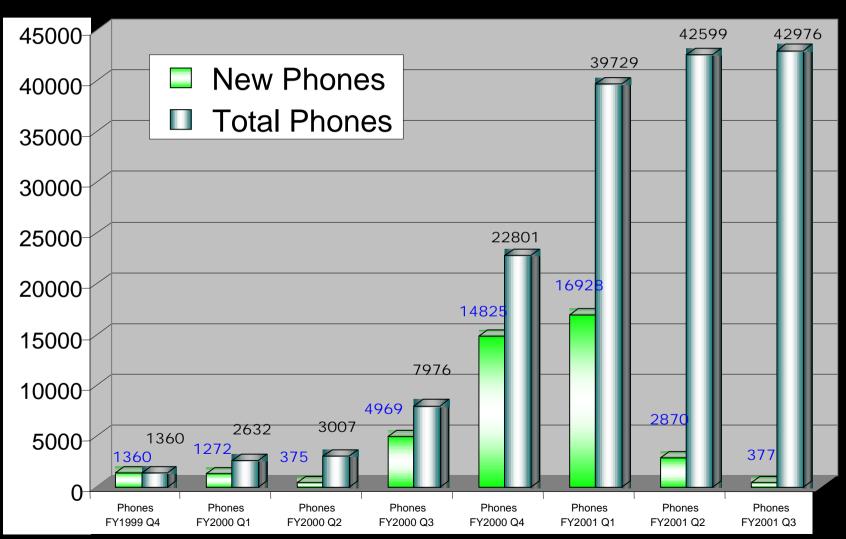
- Worldwide deployment is 5,000 IP Phones at 42 locations.
- " 32 sites are in production with 4,700 IP Phones (1)
- " 10 sites are participating in technology trials with 200 IP Phones
- The San Jose Call Manager is supporting 3,000 IP Phones in Production (2)

- (1) The TDM PBX has been removed
- (2) The IP Phone is the sole means of communication

## Cisco worldwide



### IP Phones Deployed in Cisco by Quarter



# **IP Telephony Deployment**

Theater	FY1999 Q4	FY2000 Q1	FY2000 Q2	FY2000 Q3	FY2000 Q4	FY2001 Q1	FY2001 Q2	FY2001 Q3	FY2001 Q4	FY2002 Q1-Q3
Asia Pacific	3	8	3	2	4	2	3	3	5	5
America's	0	0	2	0	8	20	20	21	21	20
Corporate	1	0	1	0	5	2	1	3	0	0
EMEA	8	10	4	0	11	6	9	0	0	0
New (Incremental) Sites	12	18	10	2	28	30	33	27	26	25
Total	12	30	40	42	70	100	133	160	186	211

## NZ Government Department

- " Legacy PBX lease expiring
- " Phase 1: 500 7960
- " Phase 2: 7000 7910
- " uOne voice mail
- " 210 sites

## Texas Instruments



- 5 sites in USA linked with ATM OC3
- Philippines and India with 512K and 256K links
- " 1000 users registering with one Call Manager cluster
- On-going deployment of AVVID





## Why AVVID?

- " Data, voice and video are converging rapidly
- " Customers are adopting today
- " Convergence in infrastructure, applications and clients

## Reasons for convergence



Lower cost

Infrastructure consolidation

Toll bypass

Lower cost of ownership

Increased productivity

Rapid application enablement

Flexibility/mobility

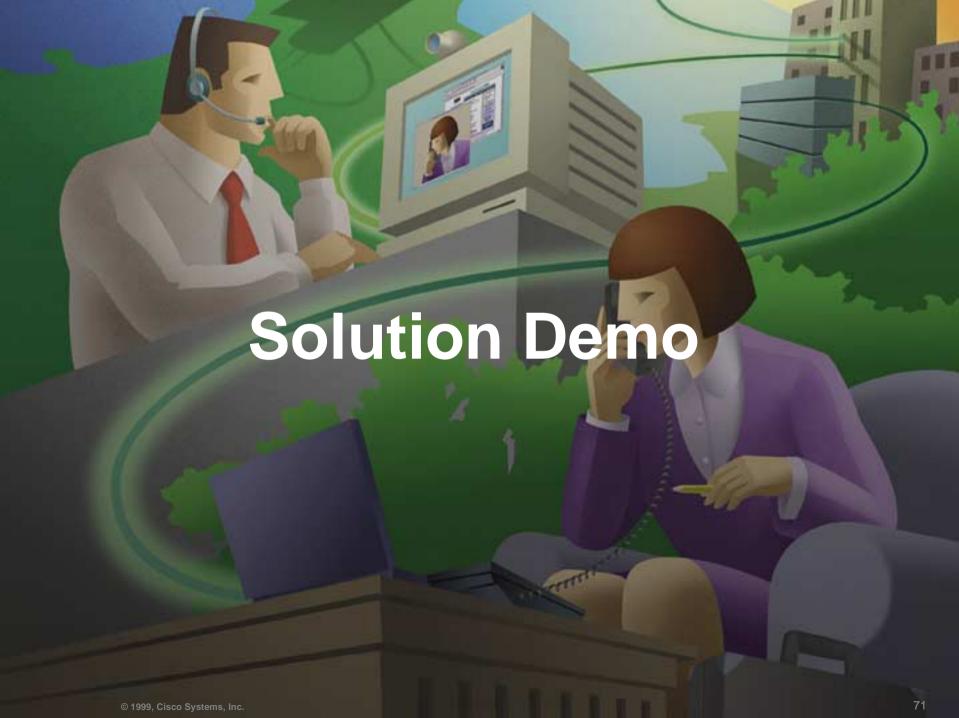
Open standards based

Competitive advantage

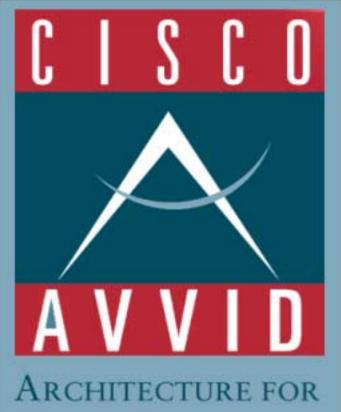


## Key Messages

- " AVVID is a total Architecture
- " Cisco has the leading deliverable solutions
- " AVVID is available today!

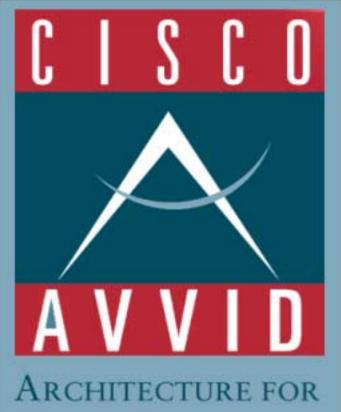






ARCHITECTURE FOR VOICE, VIDEO AND INTEGRATED DATA





ARCHITECTURE FOR VOICE, VIDEO AND INTEGRATED DATA

