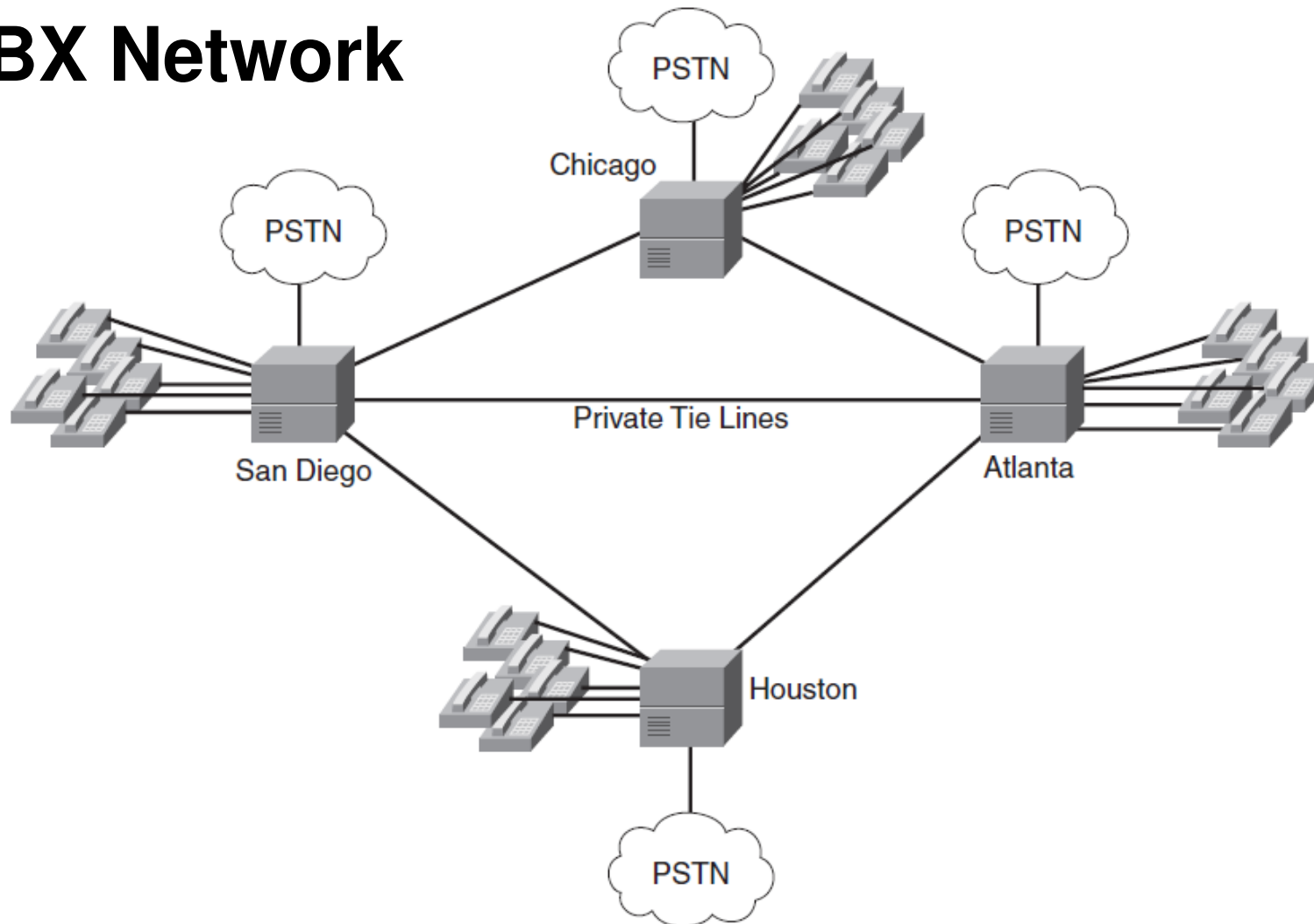




Chapter 15

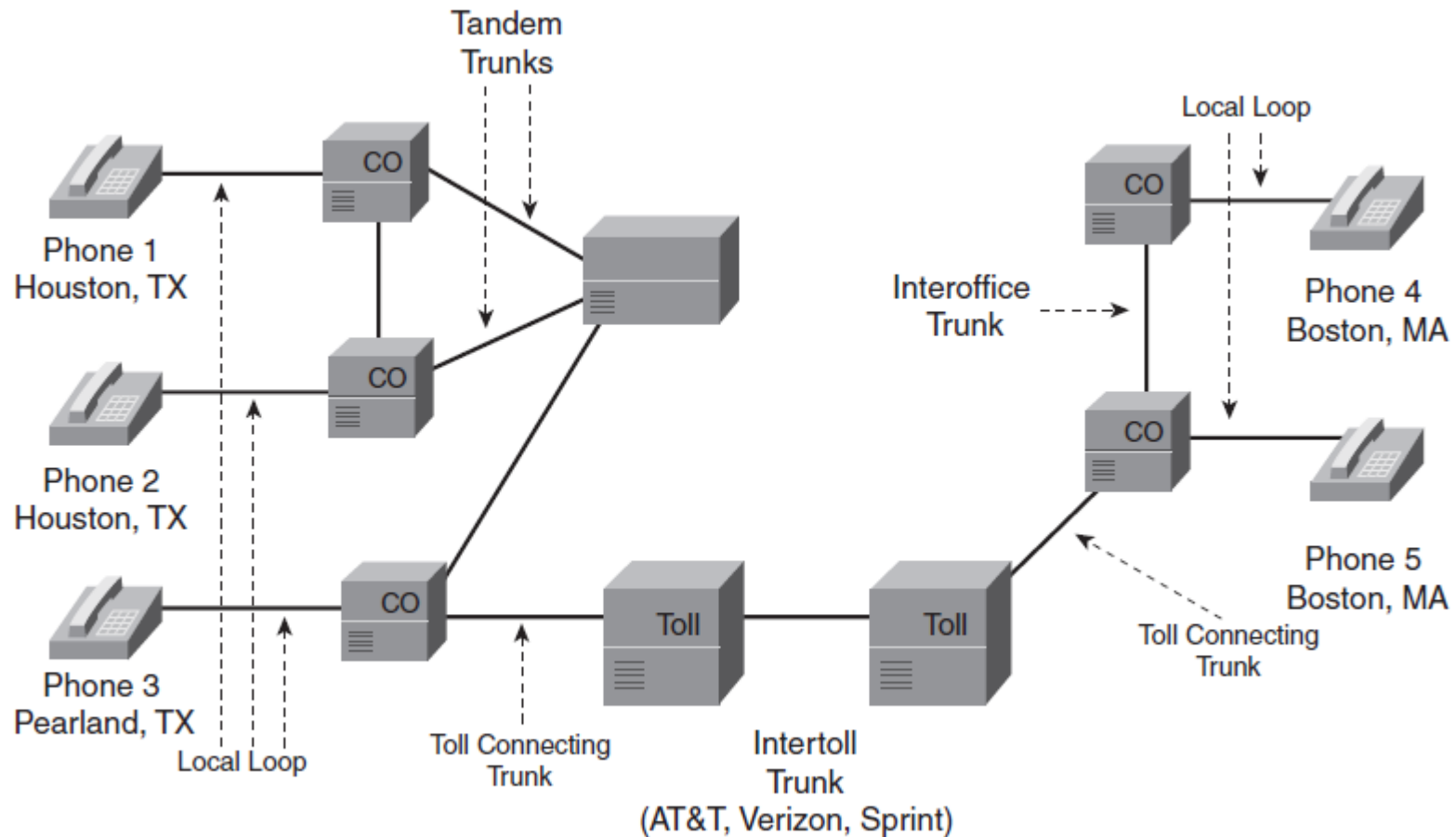
PBX Network





Chapter 15

Trunks



Chapter 15

PSTN

•Ports

- Foreign Exchange Station (FXS)
- Foreign Exchange Office (FXO)
- Ear and Mouth (E&M)
- Channelized T1 (or E1)
- ISDN Primary Rate Interface (PRI)

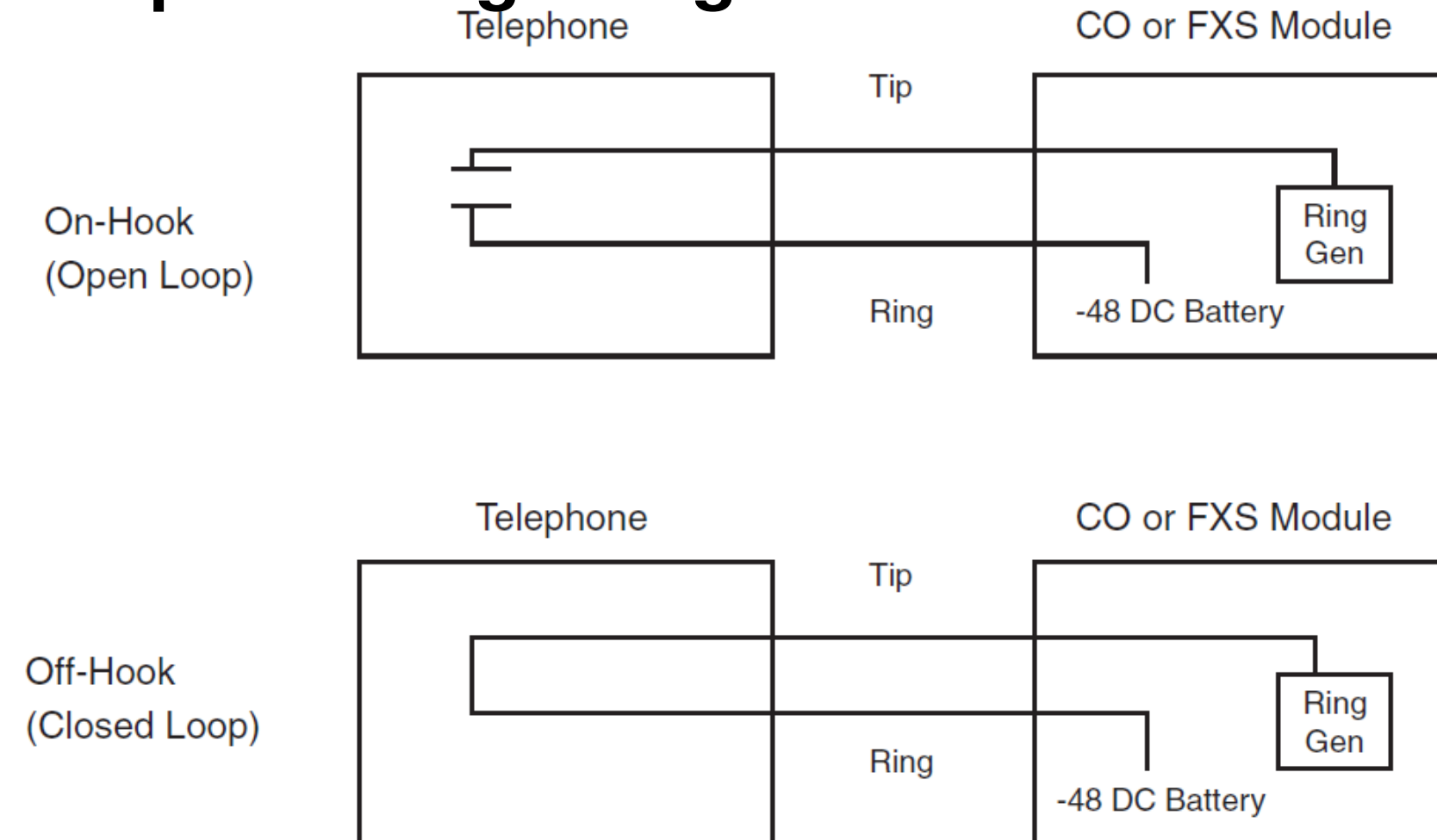
•Signaling Types

- Supervisory** provides call control and phone state (on-hook and off-hook)
- Addressing** provides dialed digits
- Informational** provides information such as dial and busy tones



Chapter 15

Loop-Start Signaling



Chapter 15

E&M Signaling

- Analog
- PBX-to-PBX

- Immediate start – Venter 200ms og sender digits
- Wink start – Venter på off-hook og sender så digits
- Delay dial – Venter 200ms, if off-hook send digits else wait

Chapter 15

CAS and CCS Signaling

- CAS bruger in-band signaling
 - T1 or E1
 - DTMF
- CCS bruger out-of-band signaling
 - ISDN PRI or BRI

Chapter 15

Addressing Digit Signaling

- Pulse or rotary dialing
- Dual-tone multifrequency (DTMF) dialing

Frequency	1209 Hz	1336 Hz	1477 Hz
697 Hz	1	ABC 2	DEF 3
770 Hz	GHI 4	JKL 5	MNO 6
852 Hz	PRS 7	TUV 8	WXY 9
941 Hz	*	OPER 0	#

Chapter 15

Other PSTN Services

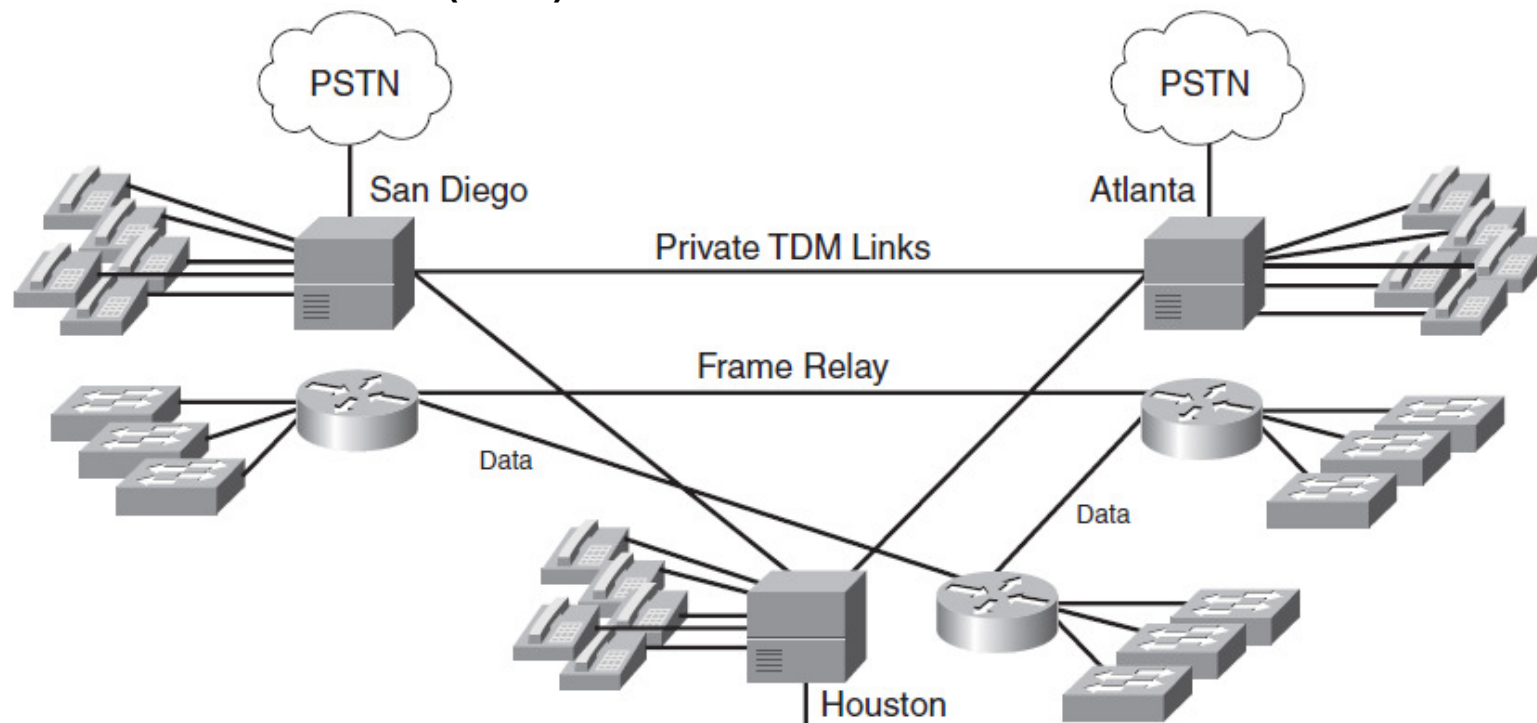
- Centrex
- Voice mail
- Database services
- Interactive voice response (IVR)
- Automatic call distribution (ACD)

- Erlang

Chapter 15

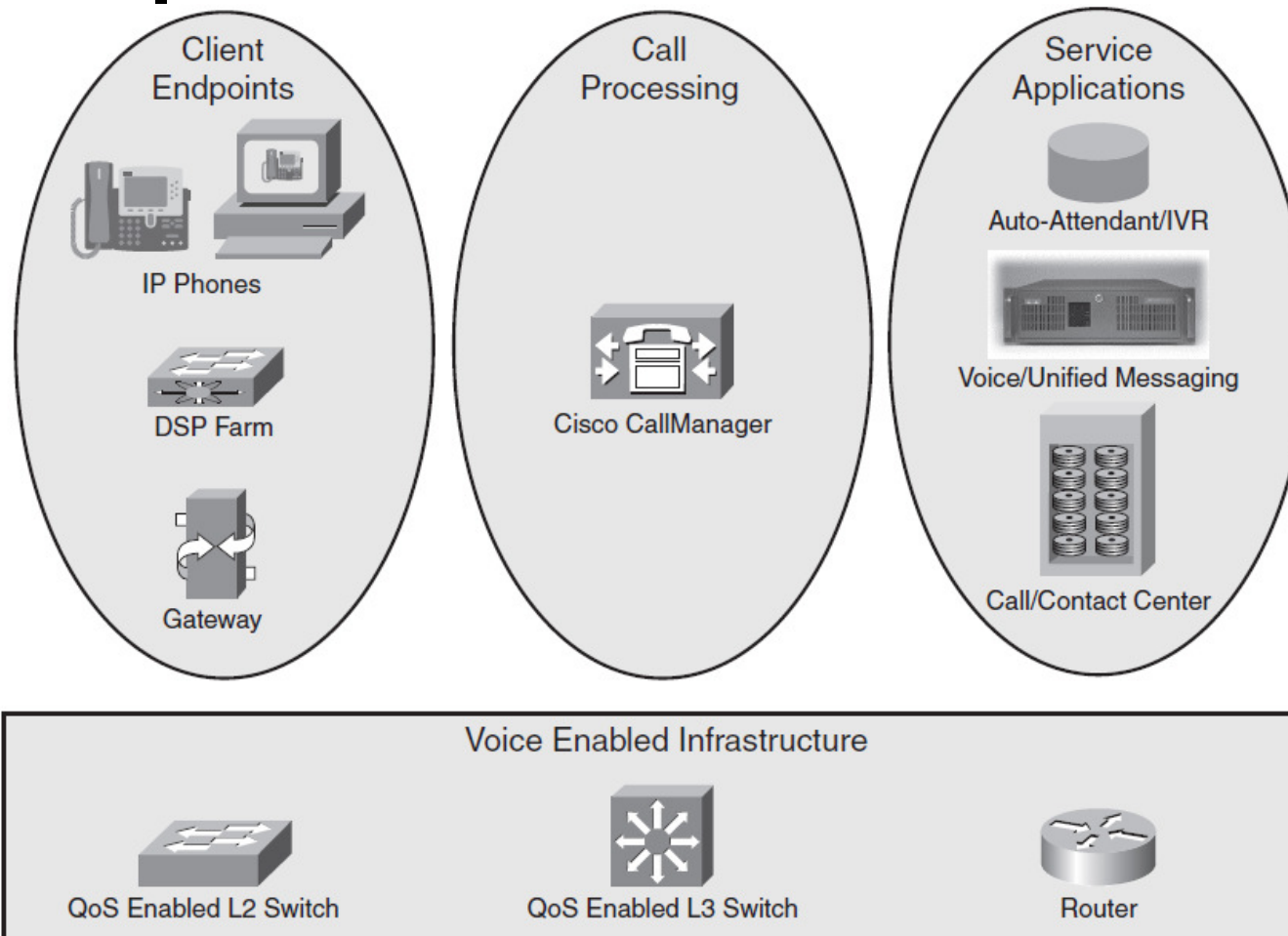
Integrated Multiservice Networks

- Voice over Frame Relay (VoFR)
- Voice over Asynchronous Transfer Mode (VoATM)
- Voice over Internet Protocol (VoIP)



Chapter 15

IPT Components



Chapter 15

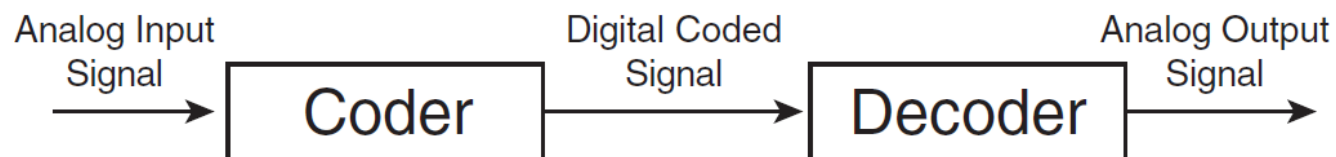
IPT Deployment Models

- **Single-Site Deployment**
- **Multisite centralized WAN call processing**
- **Multisite distributed WAN call processing**
- **CallManager Express deployment**

Chapter 15

IPT Codex

Codec	Bit Rate	MOS
G.711u	64 kbps	4.1
G.711a	64 kbps	4.1
G.723.1	6.3 kbps	3.9
G.723.1	5.3 kbps	3.65
G.726	16/24/32/40 kbps	3.85
G.728	16 kbps	3.61
G.729	8 kbps	3.92



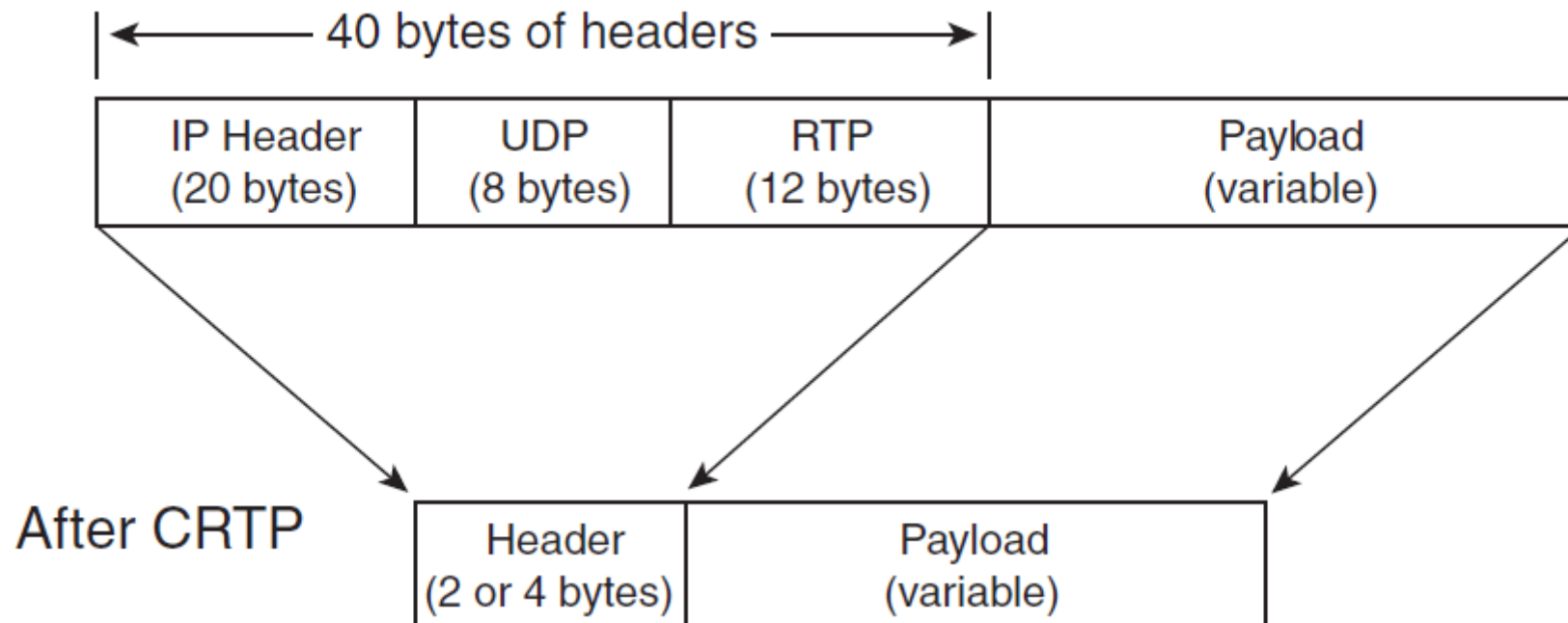
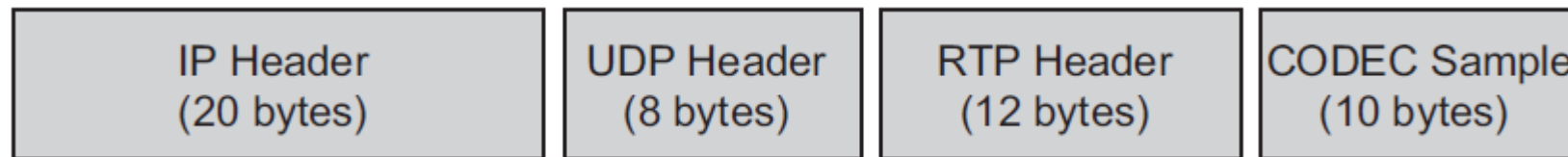
Chapter 15

VoIP Control and Transport Protocols

- Dynamic Host Configuration Protocol (DHCP)
- Domain Name System (DNS)
- TFTP
- Skinny Station Control Protocol (SSCP)
- Session Initiation Protocol (SIP)
- Real-time Transport Protocol (RTP)
- Real-time Transport Control Protocol (RTCP)
- Media Gateway Control Protocol (MGCP)
- H.323

Chapter 15

Compressed RTP (CRTP)



Chapter 15

H.323 Protocols

	Video	Audio	Data	Transport
H.323 protocol	H.261	G.711	T.122	RTP
	H.263	G.722	T.124	H.225
		G.723.1	T.125	H.235
		G.728	T.126	H.245
		G.729	T.127	H.450.1
			H.450.2	
			H.450.3	
			X.224.0	

Chapter 15

IPT Design

- **Bandwidth**
- **Voice Activity Detection (VAD)**
- **Delay**
 - Propagation delay
 - Processing delay
 - Serialization delay
 - Dejitter delay
- **QoS**
 - CRTP
 - LFI - Link Fragmentation and Interleaving
 - PQ-WFQ
 - LLQ
 - Auto QoS

Chapter 15

Auto QoS

•WAN

- Automatically classifies RTP and VoIP control packets
- Builds VoIP Modular QoS in the Cisco IOS Software
- Provides LLQ for VoIP bearer traffic
- Provides minimum-bandwidth guarantees by using CBWFQ for VoIP control traffic
- Enables WAN traffic shaping where required
- Enables LFI and RTP where required

•LAN

- Enforces a trust boundary at the Cisco IP Phone
- Enforces a trust boundary on the Catalyst switch access and uplink and downlink ports
- Enables strict priority queuing and weighted round robin for voice and data traffic
- Modifies queue admission criteria by performing CoS-to-queue mapping
- Modifies queue sizes, as well as queue weights where required
- Modifies CoS-to-DSCP and IP Precedence-to-DSCP mappings

Chapter 15

IPT Design Recommendations

- Use separate VLANs/IP subnets for IP phones.
- Use private IP addresses for IP phones.
- Place CallManager and Unity servers on filtered VLAN/IP subnets in the server access in the data center.
- Use IP precedence or DSCP for classification and marking.
- Use LLQ on WAN links.
- Use LFI on slower-speed WAN links.
- Use CAC to avoid oversubscription of priority queues.

Chapter 15



?