TCP/IP Protocol Architecture

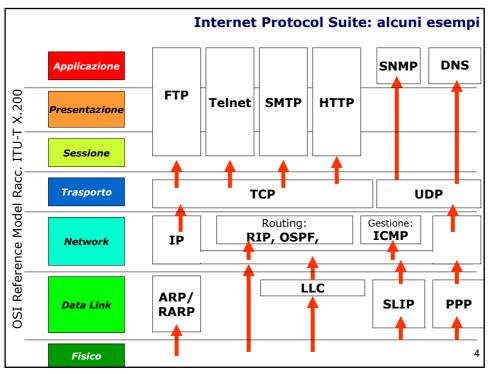
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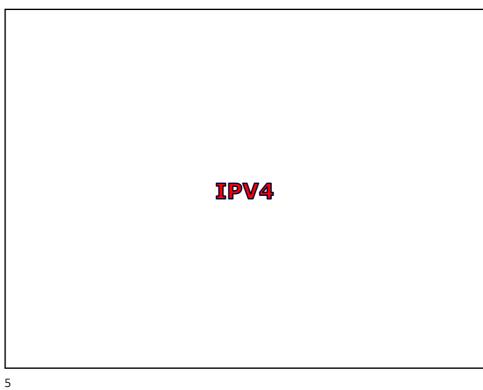
Overview

- •Internet Protocol Suite
- IPv4
- ICMP
- UDP
- TCP

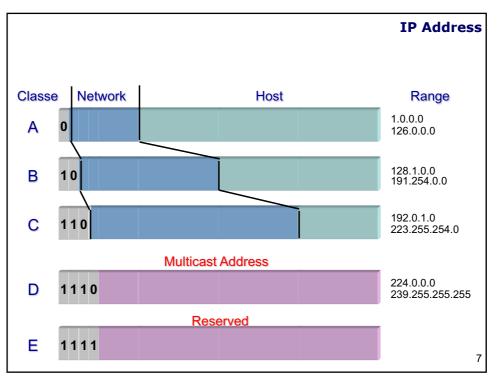
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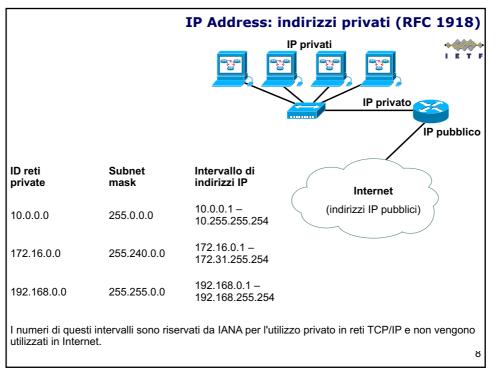
INTERNET PROTOCOL SUITE

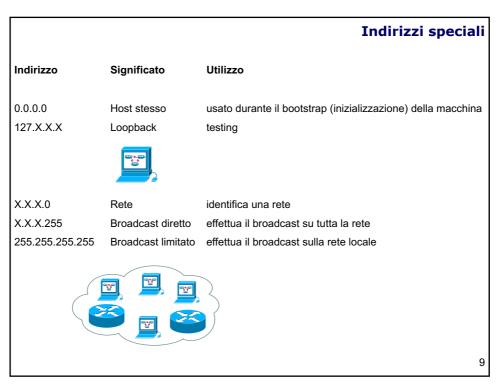




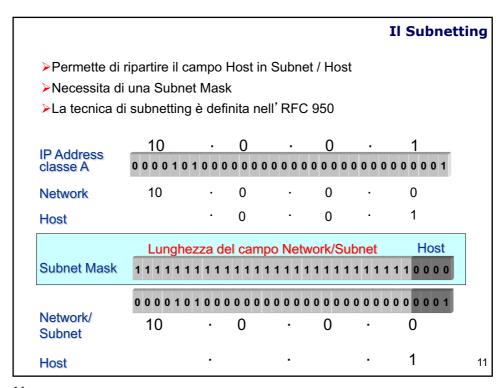
					Р	roto	ocollo IF
0 4	8	16				31	Bit
Vers. II	HL S	Service Type	Total Lenght				
Identification		Flags	Fra	gment Offset			
TTL		Protocol	Header Checksum				
Source IP Address							
Destination IP Address							
Options				Pad			
protocollo superiore (es. TCP) Lunghezza max 64KByte							

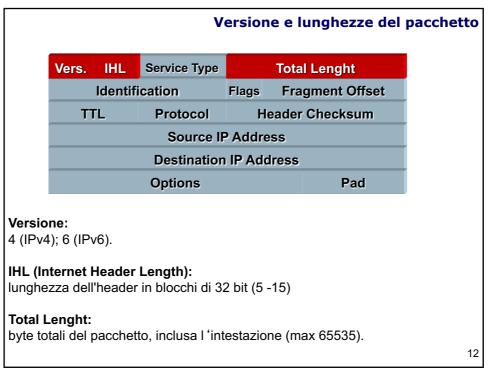


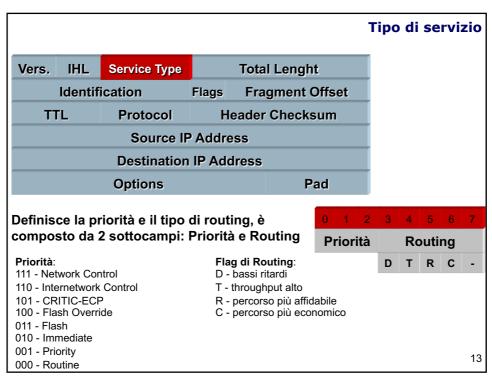


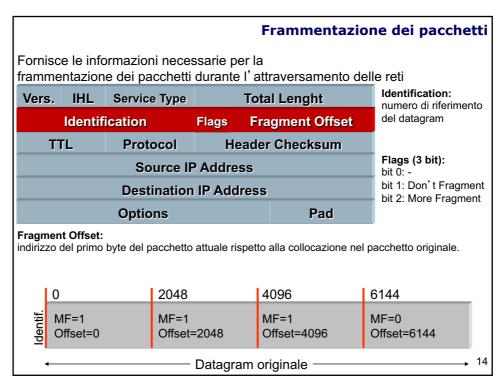


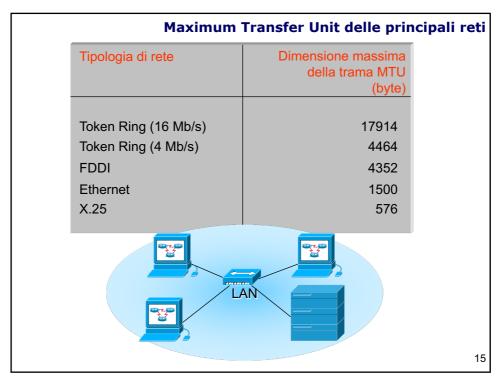
	Indirizzi IP multicast
	Alcuni indirizzi IP multicast
224.0.0.1 224.0.0.2	All Systems on this Subnet All Routers on this Subnet
224.0.0.11 224.0.0.12	Mobile-Agents DHCP Server / Relay Agent
224.0.0.18	VRRP
224.0.1.1	NTP - Network Time Protocol
224.0.1.6 224.0.1.7	NSS - Name Service Server AUDIONEWS - Multicast
224.0.1.10 224.0.1.11 224.0.1.12 224.0.1.13 224.0.1.14 224.0.1.15 224.0.1.16	IETF-1-LOW-AUDIO IETF-1-AUDIO IETF-1-VIDEO IETF-2-LOW-AUDIO IETF-2-AUDIO IETF-2-VIDEO MUSIC-SERVICE
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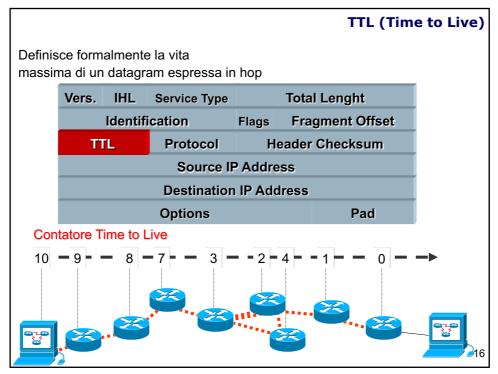


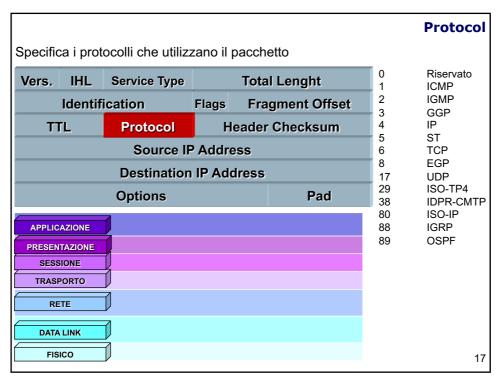


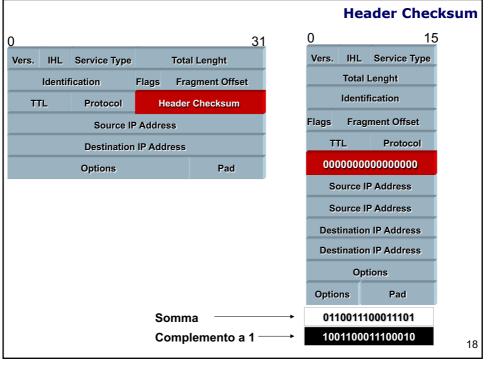


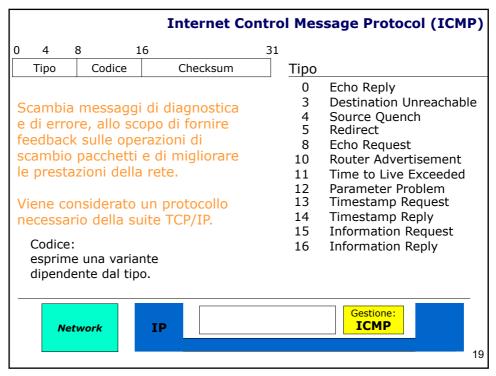


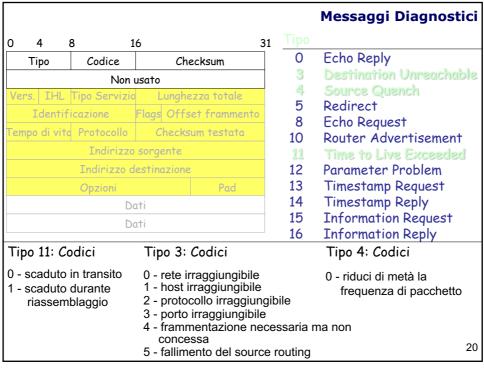




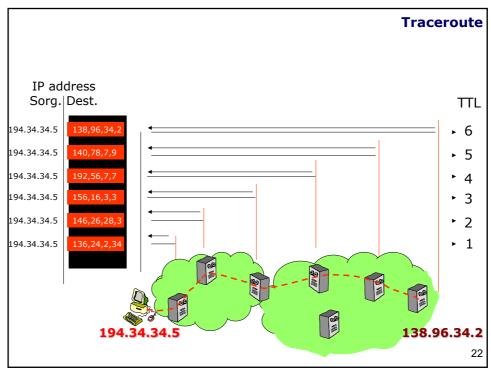


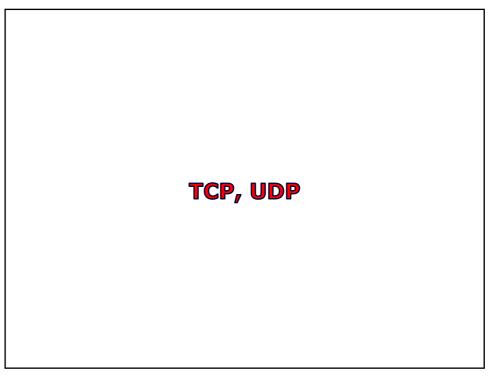


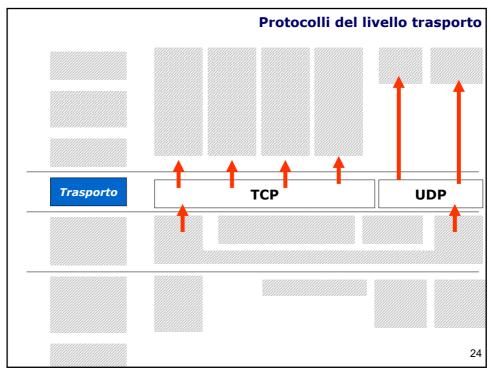




Tipo Codice Checksum Identificatore Numero sequenza Dati		
Dati Dati		
108 bytes from 198.45.45.10: icmp-seq=16. time=152. ms 108 bytes from 198.45.45.10: icmp-seq=17. time=199. ms 108 bytes from 198.45.45.10: icmp-seq=18. time=220. ms	3 4 5 10 11 12 13 14 15 16	Destination Unreachable Source Quench Redirect Echo Request Router Advertisement Time to Live Exceeded Parameter Problem Timestamp Request Timestamp Reply Information Request Information Reply







Il Protocollo TCP (Transport Control Protocol)

- TCP (Transport Control Protocol) è un protocollo di livello 4 (trasporto)
- Definito da RFC 1122/1123 ... e decine di altri!
- È un protocollo:
 - ✓ Full-duplex
 - ✓ Connection-oriented
 - ✓ Garantisce consegna affidabile ed in sequenza
 - ✓ Controllo velocità di emissione dei dati

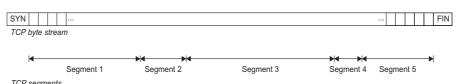
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Il Protocollo TCP

- Ritrasmette se non riceve conferma di ricezione
- Esegue controllo di congestione end-to-end per evitare che la rete venga utilizzata oltre la sua capacità
- Esegue il controllo di flusso end-to-end perché un host veloce non saturi un host lento
- Frammenta (o raccoglie) l'informazione in segmenti di dimensione opportuna
- Mette in sequenza i datagram IP che arrivano fuori sequenza

A Segmented TCP Stream

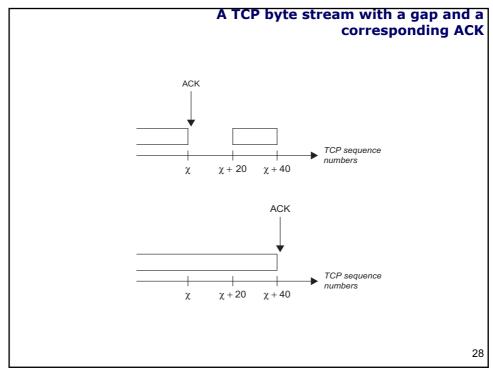


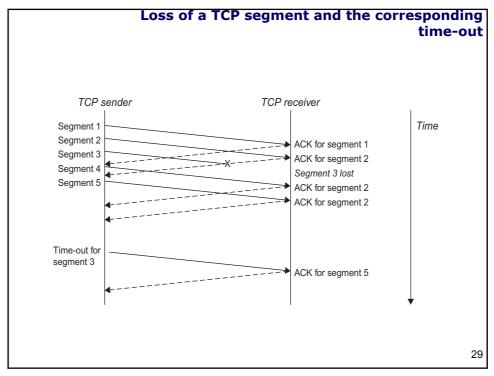
TCP segments

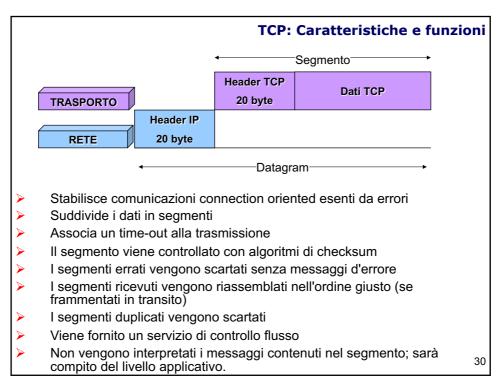
- Each byte in the TCP byte stream is assigned a sequence number.
- The stream is partitioned into segments that may be arbitrarily sized.
- Each segment is prepended with a TCP header and transmitted in separate IP packets.
- In theory, for each received segment the receiver produces an ACK. In practice, however, most TCP implementations send an ACK only on every other incoming segment to reduce ACK traffic.
- ACKs are also piggybacked on outgoing TCP segments.
 - The ACK contains the next sequence number expected in the continuous stream of bytes.
 - Thus, ACKs do not acknowledge the reception of any individual segment, but rather acknowledge the transmission of a continuous range of bytes.

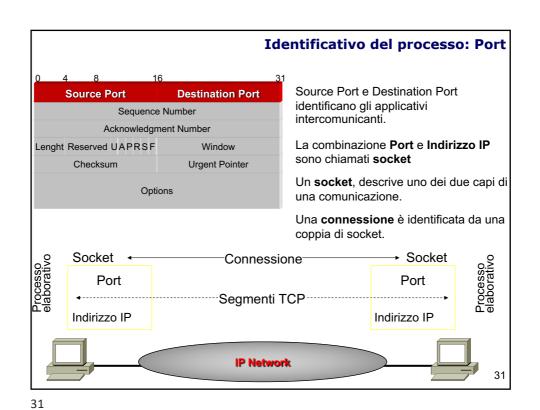
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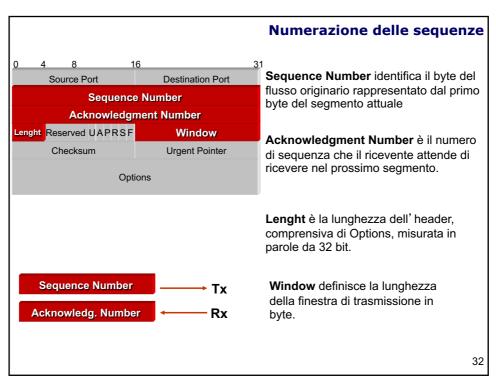
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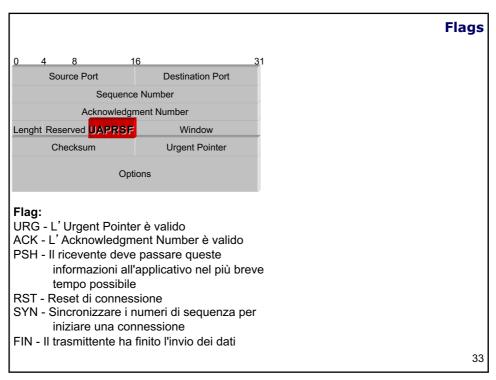


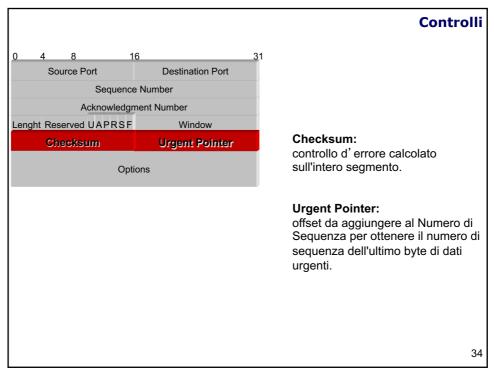


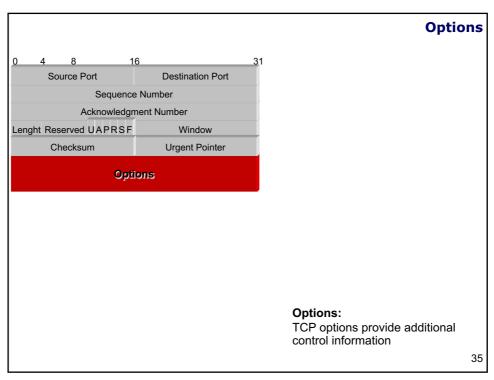


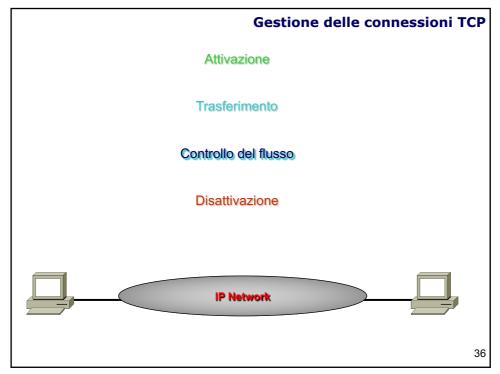


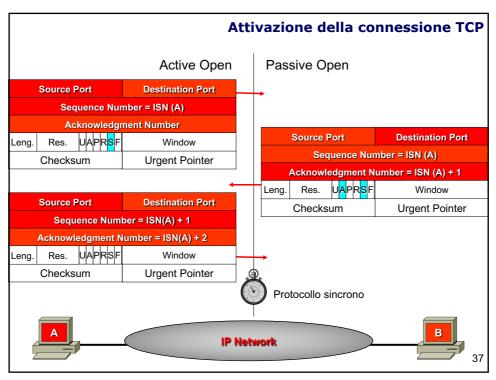


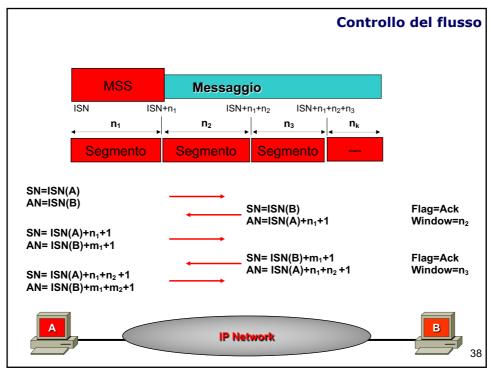


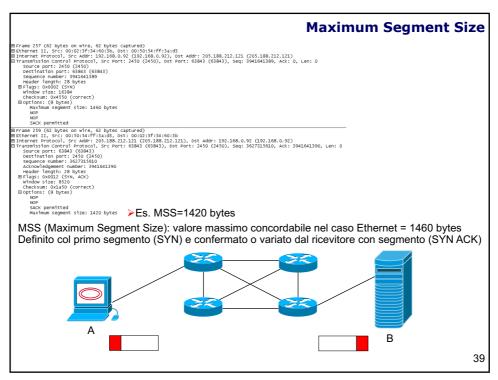


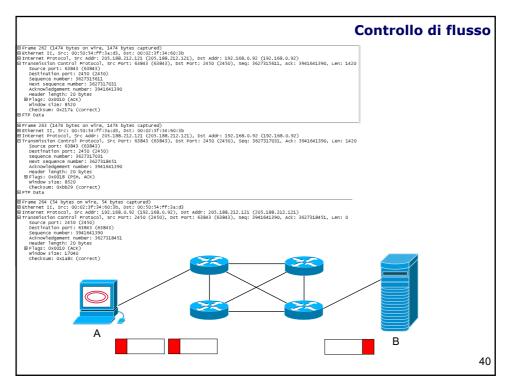


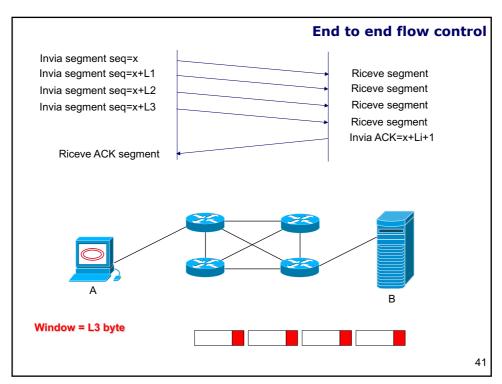


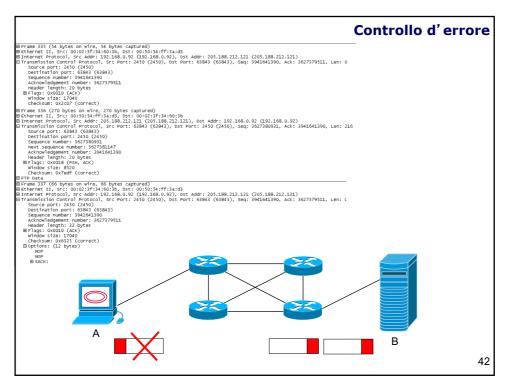


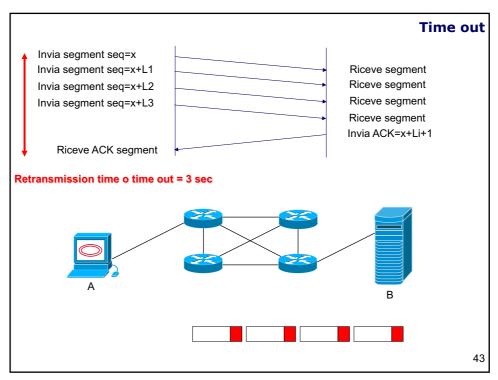


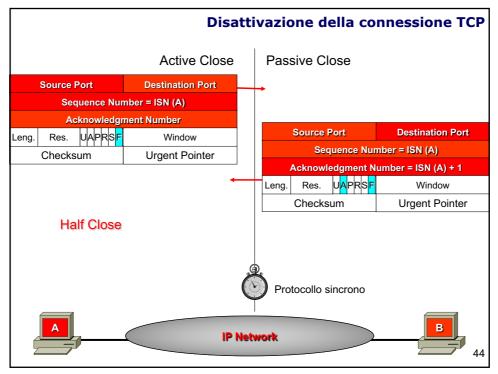












Congestion Control

If flow control ignores that the buffer space will be overrun at the end points, the congestion control mechanisms try to prevent the overrun of router buffer space.

To achieve this TCP uses two separate methods:

- •Slow start: Probes the available bandwidth when starting to send over a connection
- •Congestion avoidance: Constantly adapts the sending rate to the perceived bandwidth of the path between the sender and the receiver

For smart object networks, which may have only limited amounts of data to send, TCP congestion control is rarely invoked

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Round-Trip Time Estimation

A critical factor of any reliable protocol is the **round-trip time estimation**, since the round-trip time determines the time to wait for an ACK before retransmitting a segment

- If the round-trip time estimate is much lower than the actual round-trip time of the connection, segments will be retransmitted before the original segment or its corresponding ACK has propagated through the network
- If the round-trip time estimation is too high, time-outs will be longer than necessary, thus reducing performance

TCP uses feedback provided by its acknowledgment mechanism to measure round-trip times

Round-trip time measurements are taken once per window

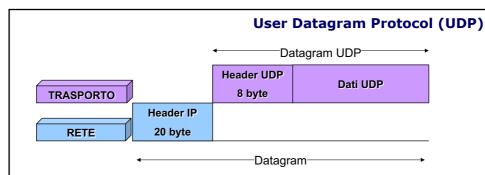
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Il Protocollo UDP (User Datagram Protocol)

- UDP (User Datagram Protocol) permette alle applicazioni di un host l'invio di datagram ad altre applicazioni di un host remoto
- Definito da RFC-768 (1980)
- UDP fornisce un servizio di livello 4, ma:
 - √Connectionless (pacchetti fuori sequenza)
 - ✓ Non affidabile (pacchetti persi)
 - ✓ Senza controllo di flusso (saturazione del ricevitore)

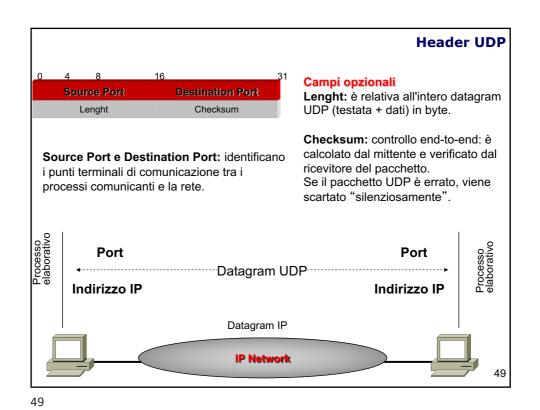
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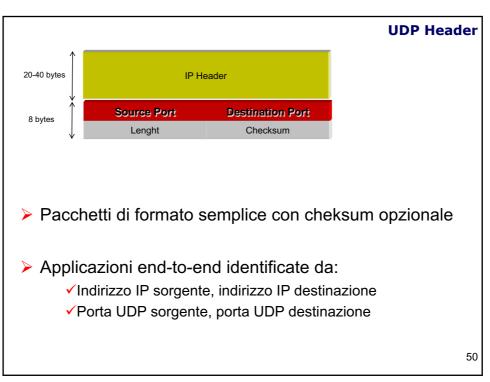
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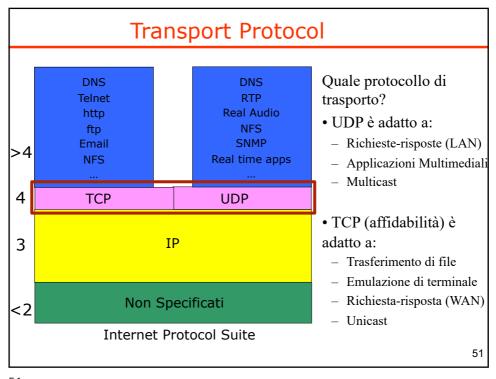


- Fornisce un servizio connection less inaffidabile
- I dati sono suddivisi in porzioni chiamate datagram UDP
- Non ci sono procedure di attivazione della connessione
- Il datagram UDP viene controllato con algoritmi di checksum solo opzionalmente
- I segmenti errati vengono scartati senza messaggi d'errore
- Non fornisce nessun servizio di controllo flusso
- Non vengono interpretati i messaggi contenuti nel segmento; sarà compito del livello applicativo.

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Protocollo di Trasporto: UDP

- Si può usare UDP?
 - UDP supporta il multicast
 - Non riordina i dati ricevuti fuori sequenza
 - Non rileva perdite
 - Non reagisce a variazioni di ritardo
 - Non identifica contenuti multimediali
- IETF: introduzione di RTP "sopra" UDP per comunicazioni multimediali

Protocollo di Trasporto: TCP

- Si può usare TCP?
 - TCP offre un trasporto affidabile, ma le ritrasmissioni ed il controllo di flusso/congestione causano:
 - Ritardi in caso di perdita
 - Variazioni della banda disponibile
 - TCP non supporta il multicast
- TCP può essere usato per trasferire "file" multimediali (in email o in pagine web)

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