

# CS2105 Lecture 4

# **Reliable Protocols**

3 February, 2014

After this class, you are expected to:

- understand the interface between the transport layer and the network layer
- be able to design your own reliable protocols with ACK, NAK, sequence numbers, timeout, and retransmission.
- know how to calculate the utilization of a channel.
- understand the working of Go-Back-N and Selective Repeat protocols

“Sending Data Reliably Over the Internet is Harder Than You Think. The Intricacy Involved in Ensuring Reliability Will Make Your Head Explode.”



Application

Transport

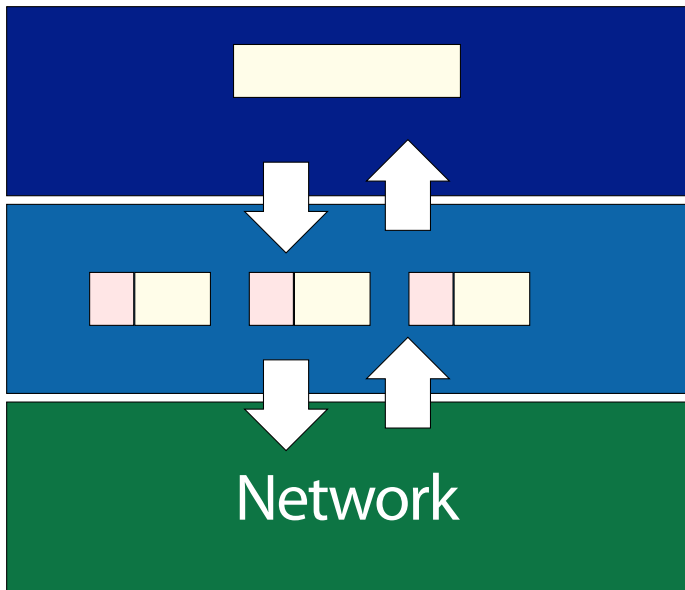
Network

Link

Physical

Transport layer resides **on end hosts** and provides **process-to-process** communication.

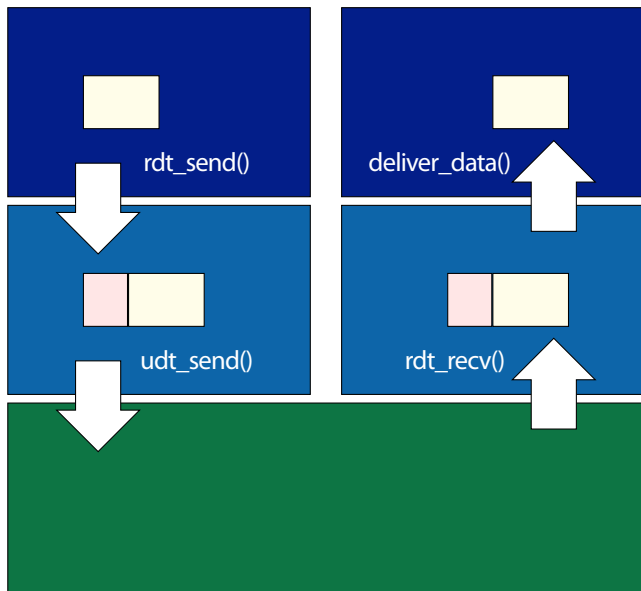
Network layer provides  
**host-to-host,**  
**best-effort,**  
**unreliable**  
communication.



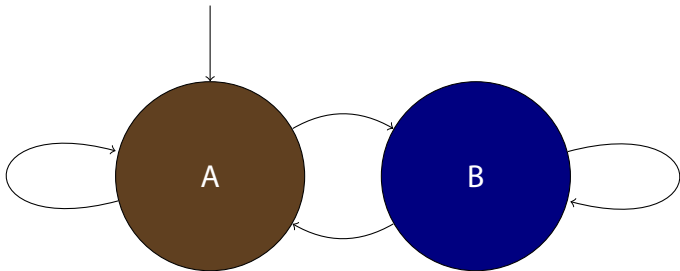
How to build a **reliable transfer protocol** on top of unreliable communication?



Unreliable: may not deliver at all or deliver with error



# Finite State Machine

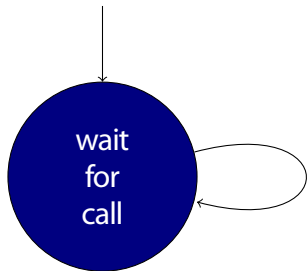


Event  
Action

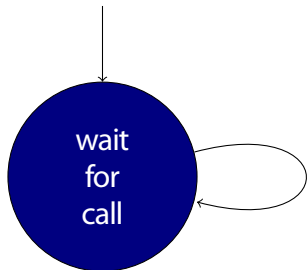
# **rdt 1.0**

Assume underlying channel is  
reliable

# rdt 1.0 sender



# rdt 1.0 receiver



## **rdt 2.0**

Underlying channel may  
introduce bit errors

# ARQ

## Automated Repeat reQuest

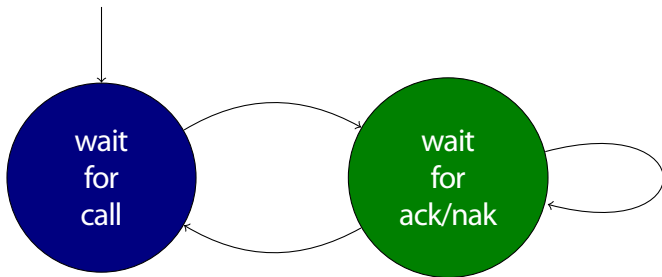


Receiver detects errors  
Receiver feeds back to sender  
Sender retransmits

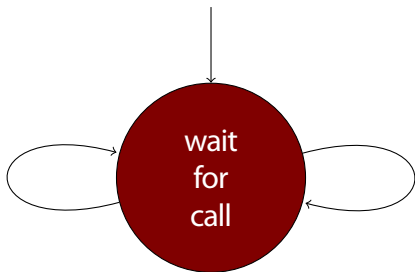
Receiver detects errors  
Receiver feeds back to sender  
Sender retransmits

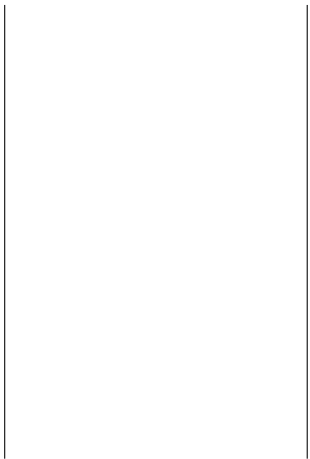
We will talk about checksum, the technique to detect errors, in more detail in the subsequent lectures.

# rdt 2.0 sender



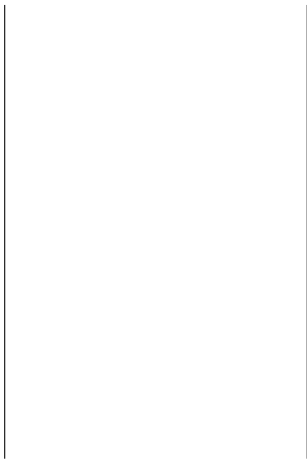
# rdt 2.0 receiver





# Stop-and-Wait Protocol

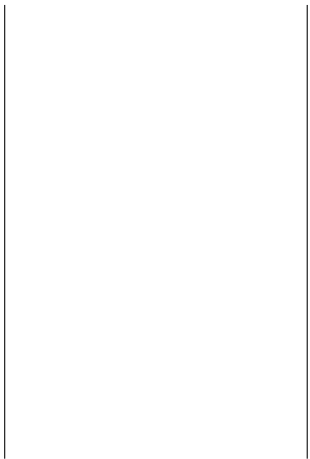
Bug:  
What if ACK/NAK is corrupted?



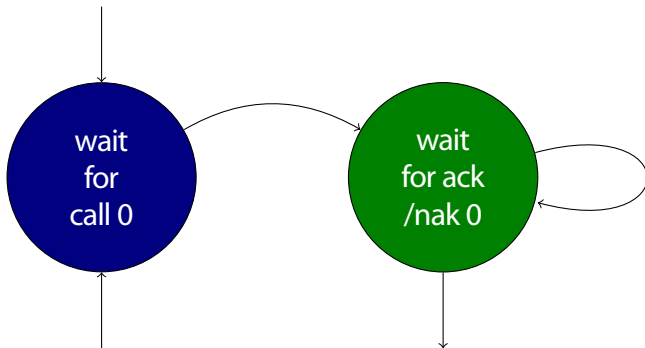


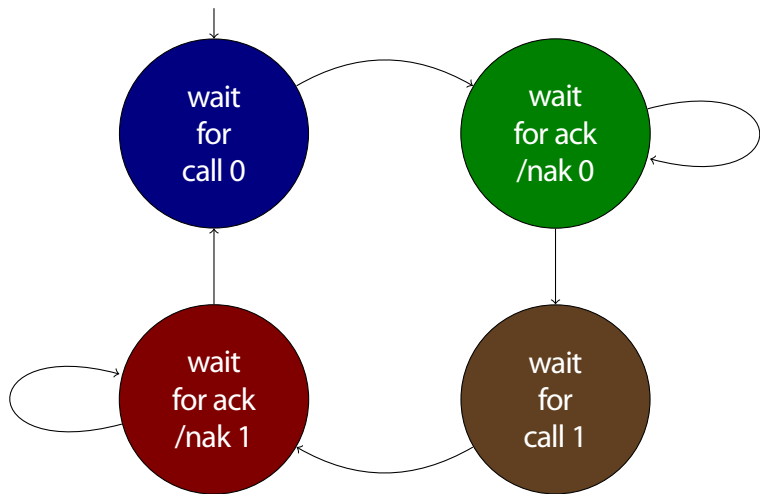
# Bug Fix:

## Add a sequence number

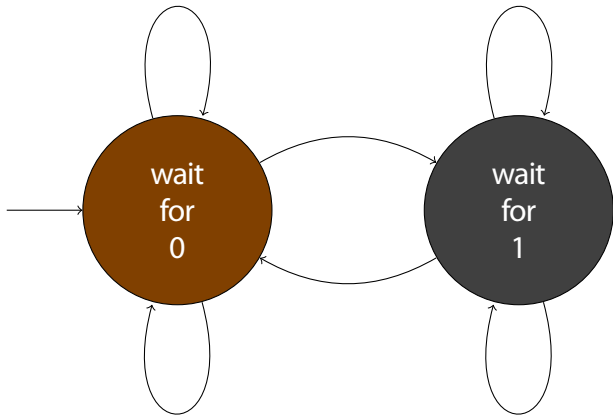


# rdt 2.1 sender





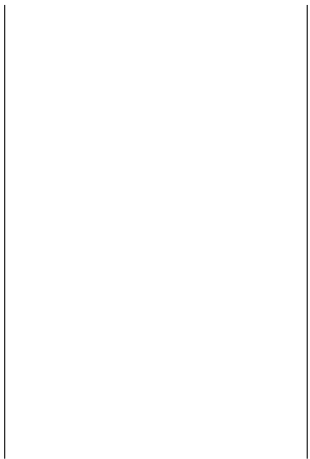
# rdt 2.1 receiver





## rdt 2.2

Replace NAK with ACK of the last correctly received packet.





## **rdt 3.0**

Packet can be loss or corrupted

Challenge: If ACK is lost, how to tell if the packet has been received?

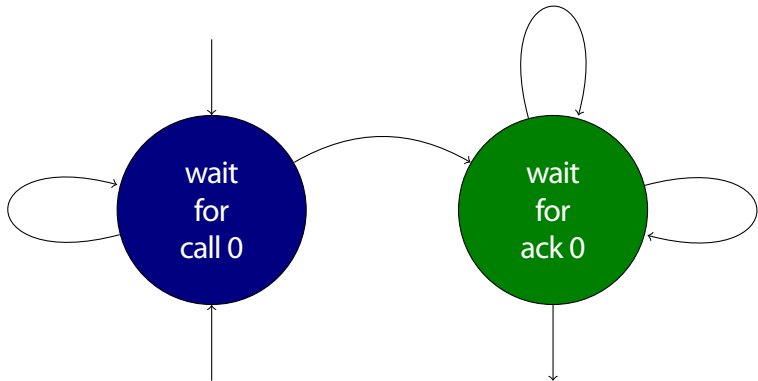
|

|

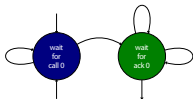
Resend after timeout  
(may lead to dups, but OK since  
we have seq numbers).



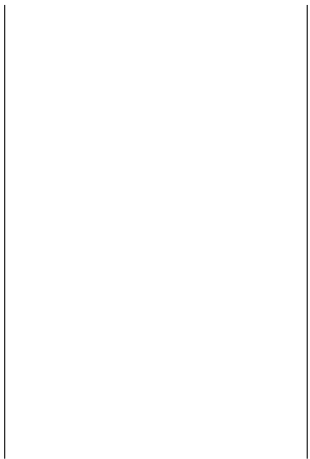
# rdt 3.0 sender



rdt 3.0 sender



The state diagram for the receiver in rdt 3.0 is given as an exercise.



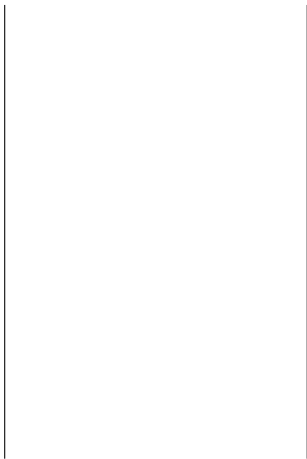


# Alternating Bit Protocol

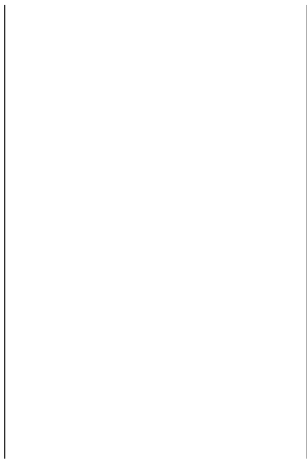
$RTT = 30\text{ms}$

$R = 1\text{ Gbps}$

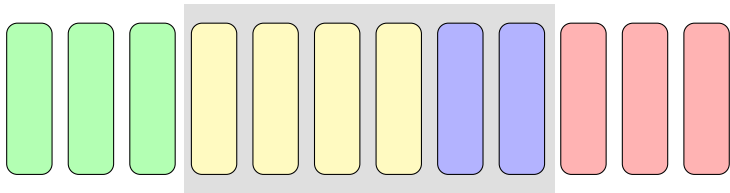
$L = 1000\text{ bytes}$

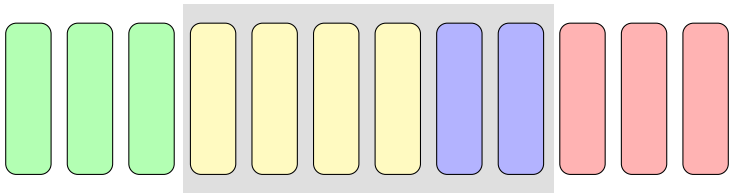


**Pipelining:**  
need buffering and larger  
range of sequence numbers.

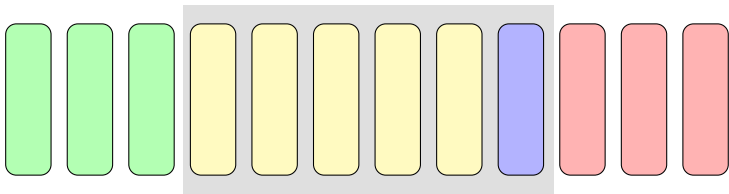


# Go-Back-N

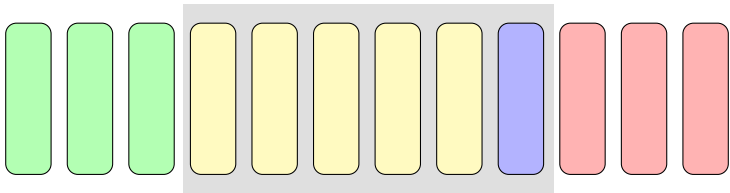




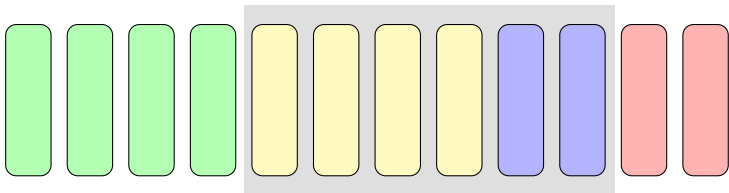
send a packet

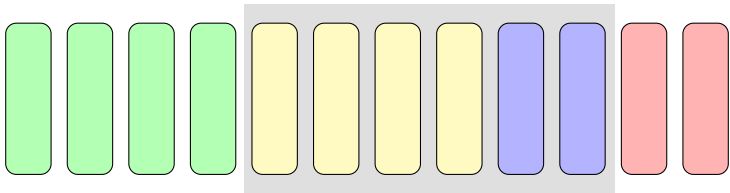




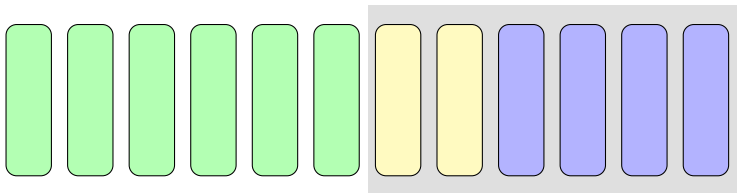


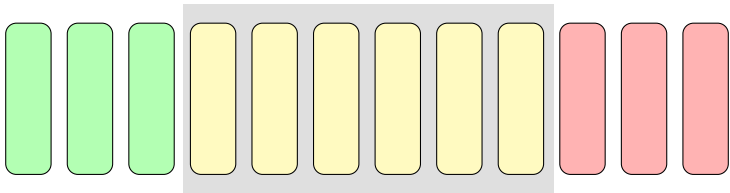
receive ACK 3



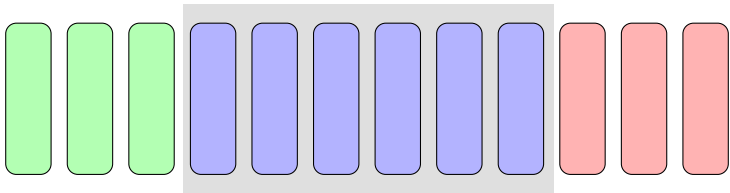


receive ACK 5





window is full



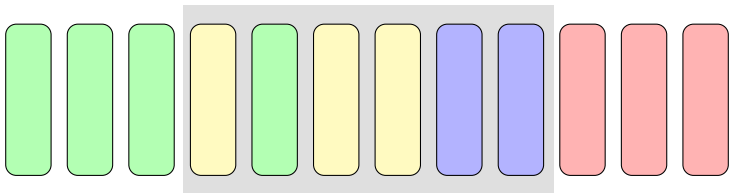
window is empty

# Selective Repeat

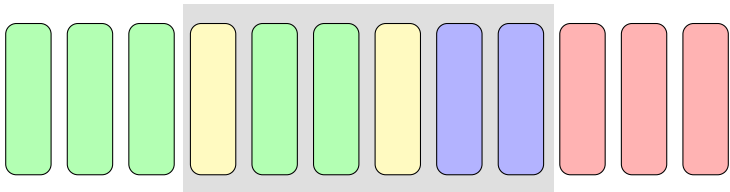


1. one timer per packet
2. receiver needs buffer

at sender:

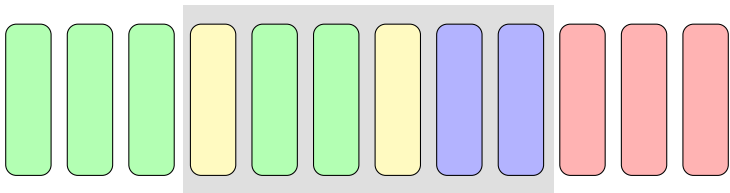


receive ACK 5

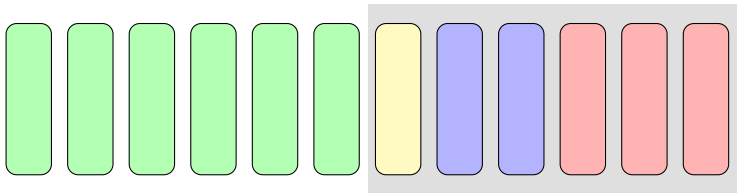




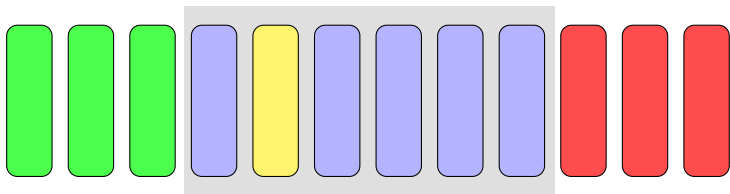
at sender:



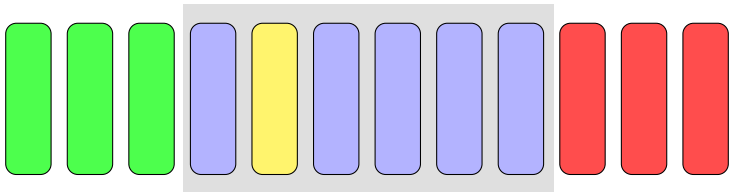
receive ACK 3



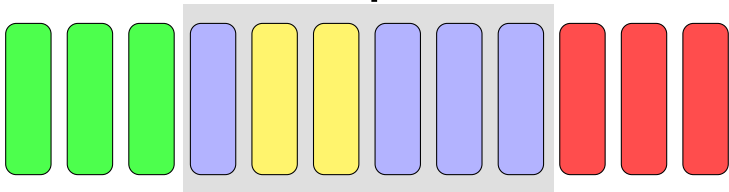
at receiver:



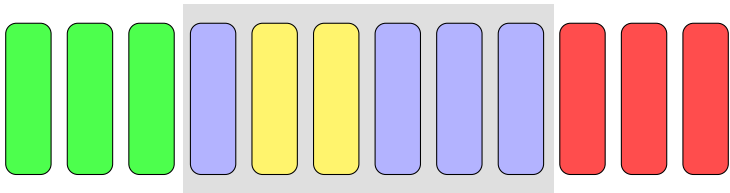
at receiver:



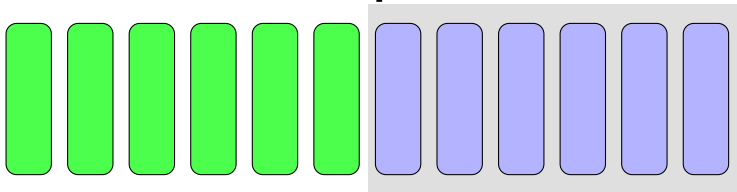
receive packet 5

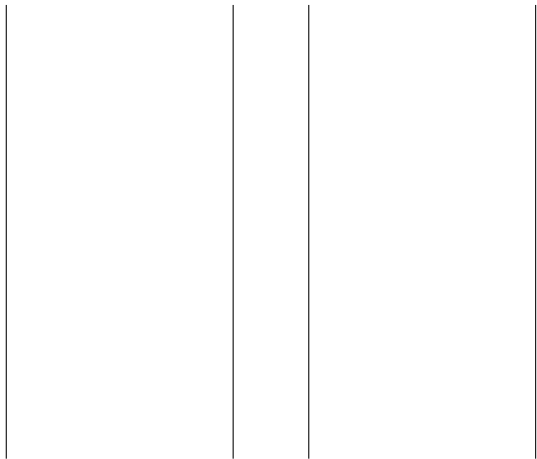


at receiver:



receive packet 3





	GBN	SR
ACK	cumulative	selective
out-of-order	ignore	keep
retransmit	all unack	one unack
timer	earliest unack	one per unack

How to build a **reliable protocol** on top of unreliable communication?

How to build a **reliable**  
**protocol** on top of unreliable  
communication?

The same techniques can be used in other layers, such as in an application layer protocol over UDP, or in a link-layer protocol.



error detection  
retransmission  
timers  
sequence numbers  
ACK/NAK  
window and pipelining