

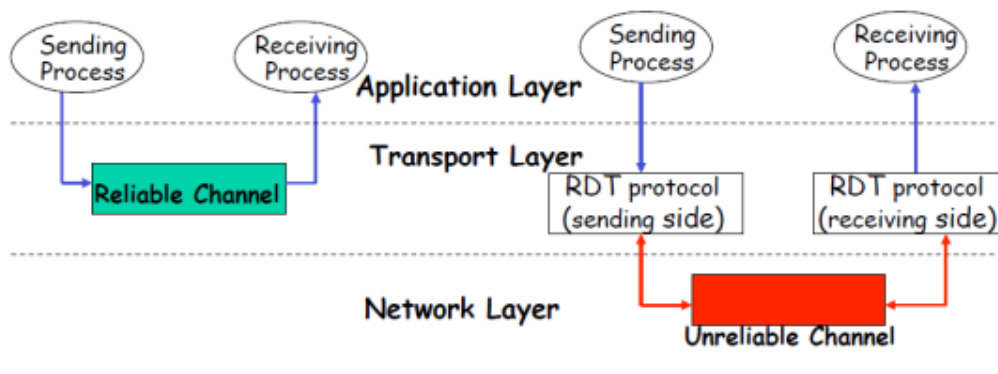


애플리케이션 계층 2

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Principles of Reliable Data Transfer

- ❑ Fundamentally important networking topic!
- ❑ Why do we need reliable data transfer protocol?



- ❑ What can happen over unreliable channel?
 - Message error
 - Message loss

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- unreliable channel
 - message error
 - message loss
 - 두가지뿐임

Let's Build *simple* Reliable Data Transfer Protocol

We'll:

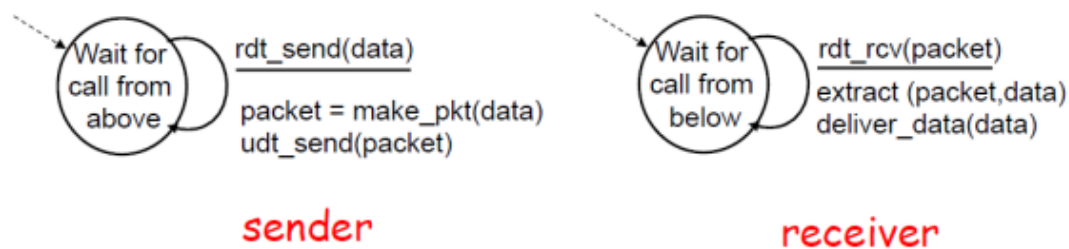
- **incrementally** develop sender, receiver sides of reliable data transfer protocol (rdt)
- consider only stop-and-wait protocol
- use finite state machines (FSM) to specify sender, receiver



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Rdt1.0: Data Transfer over a Perfect Channel

- underlying channel is perfectly reliable
 - no packet errors
 - no packet loss
- What mechanisms do we need for reliable transfer?
 - **Nothing!** Underlying channel is reliable!



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- underlying channel 완벽하면 에러 없음

Rdt2.0: channel with packet errors (no loss!)

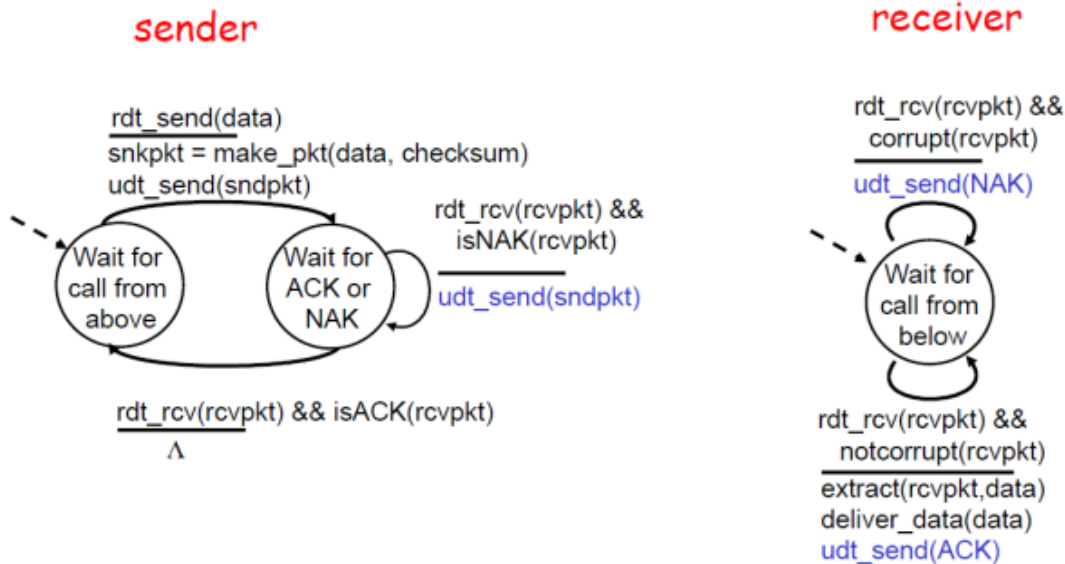
- What mechanisms do we need to deal with error?
 - Error detection
 - Add checksum bits
 - Feedback
 - *Acknowledgements (ACKs)*: receiver explicitly tells sender that packet received correctly
 - *Negative acknowledgements (NAKs)*: receiver explicitly tells sender that packet had errors
 - Retransmission
 - sender retransmits packet on receipt of NAK

- So, we need the following mechanisms:
 - Error detection, Feedback (ACK/NAK), Retransmission

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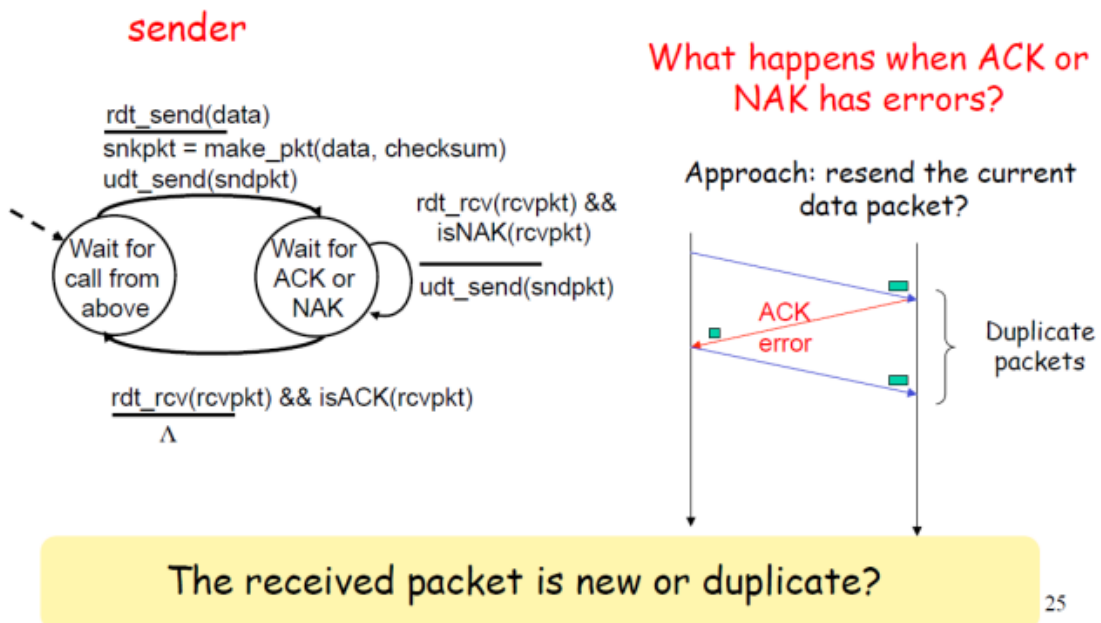
- 에러 탐지
- 지속적인 피드백
- 재전송

rdt2.0: FSM specification



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rdt2.0: Can this completely solve errors?



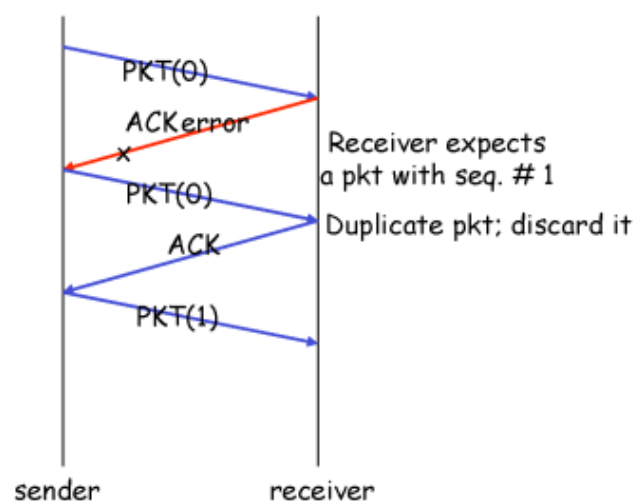
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Handling Duplicate Packets

- ❑ Sender adds *sequence number* to each packet
- ❑ Sender retransmits current packet if ACK/NAK garbled
- ❑ Receiver discards duplicate packet

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rtd2.1: examples



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- Seq# 는 header에 들어감
- header는 최소한의 필드만 사용해야함

rdt2.1: summary for packet error

- Mechanisms for channel with packet errors
 - Error detection, Feedback, Retransmission, Sequence#

Sender:

- seq # added to pkt
- must check if received ACK/NAK corrupted
- Retransmit on NAK or corrupted feedback

Receiver:

- must check if received packet is duplicate
- send NAK if received packet is corrupted
 - Send ACK otherwise

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rdt2.2: a NAK-free protocol

- ❑ Same functionality as rdt2.1, using ACKs only
- ❑ Instead of NAK, receiver sends ACK for last correctly received packet
 - Receiver must *explicitly* include seq # of pkt being ACKed
- ❑ Duplicate ACK at sender results in same action as NAK: *retransmit current pkt*

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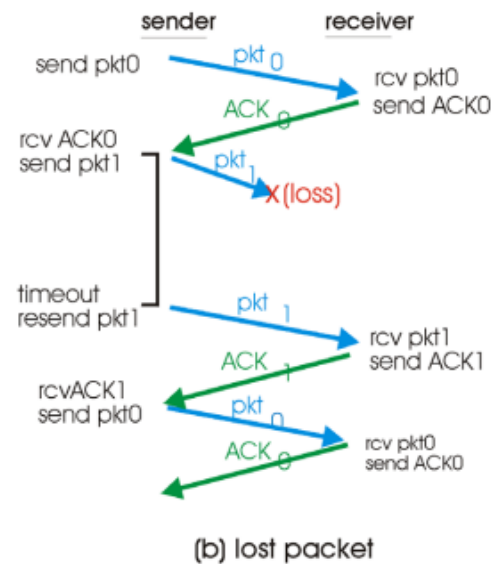
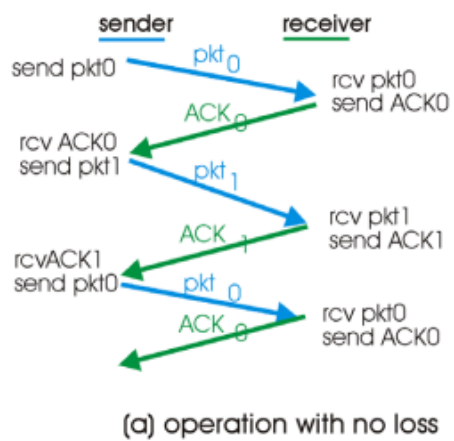
rdt3.0: channel with loss & packet errors

- ❑ What mechanisms do we need for packet loss?
 - **Timer!**
- ❑ Sender waits "reasonable" amount of time for ACK (a Time-Out)
- ❑ If packet (or ACK) is just delayed (not lost):
 - Retransmission will be duplicate, but use of seq. #'s already handles this

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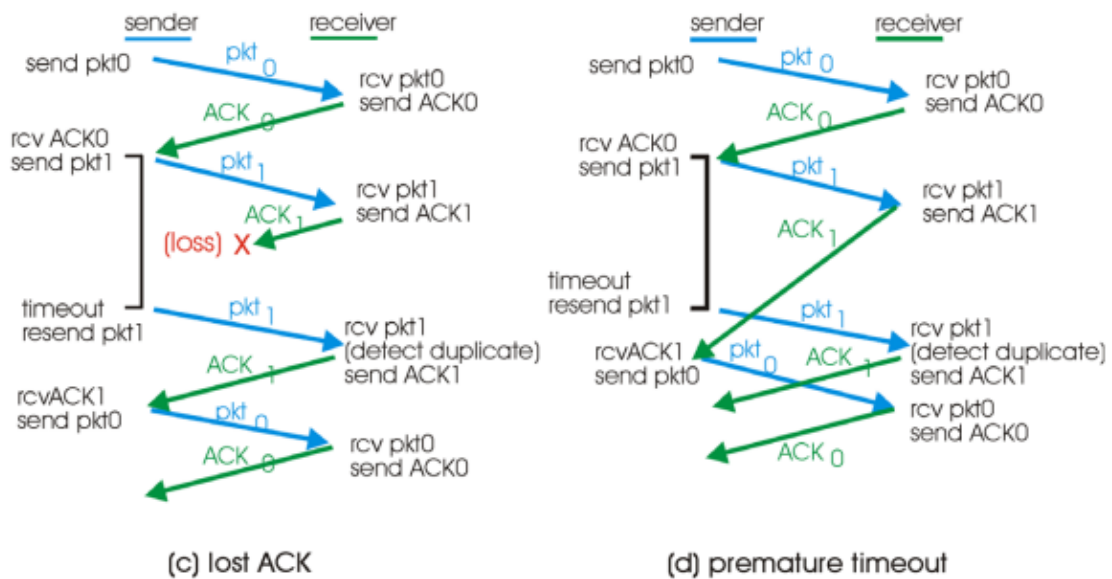
- 시간이 지나면 loss를 알게됨
- 센터는 패키지 보낼 때 마다 타이머를 같이 보냄

rdt3.0 in action



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rdt3.0 in action



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Recap: Principles of Reliable Data Transfer

- ❑ What can happen over unreliable channel?
 - Packet error, packet loss
- ❑ What mechanisms for **packet error**?
 - Error detection, feedback, retransmission, sequence#
- ❑ What mechanisms for **packet loss**?
 - Timeout!
- ❑ We built simple reliable data transfer protocol
 - Real-world protocol (e.g., TCP) is more complex, but with same principles!

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Performance of rdt3.0

- ❑ rdt3.0 works, but performance stinks
- ❑ example: 1 Gbps link, 15 ms e-e prop. delay, 1KB packet:

$$T_{\text{transmit}} = \frac{L \text{ (packet length in bits)}}{R \text{ (transmission rate, bps)}} = \frac{8\text{kb/pkt}}{10^9 \text{ b/sec}} = 8 \text{ microsec}$$

- U_{sender} : **utilization** – fraction of time sender busy sending

$$U_{\text{sender}} = \frac{L / R}{RTT + L / R} = \frac{.008}{30.008} = 0.00027$$

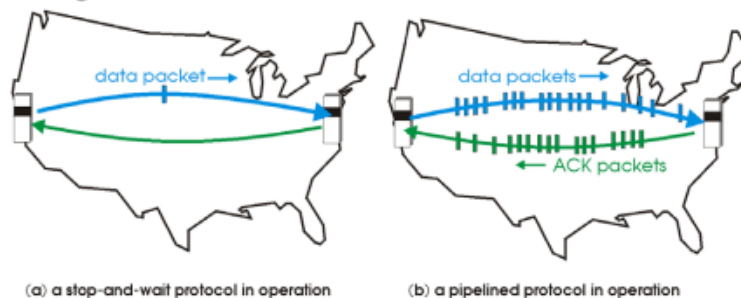
- 1KB pkt every 30 msec -> 33kB/sec thruput over 1 Gbps link
- network protocol limits use of physical resources!

Transport Layer 3-38

Pipelined protocols

Pipelining: sender allows multiple, “in-flight”, yet-to-be-acknowledged pkts

- range of sequence numbers must be increased
- buffering at sender and/or receiver



- ❑ Two generic forms of pipelined protocols: *go-Back-N*, *selective repeat*

Transport Layer 3-40

loss - timer

메카니즘은 구현되어있음

packet의 tcp 헤더부분에 정보 필드

하지만 현재 배운 것은 RDT protocol은 너무 단순해서 한번에 하나의 패킷밖에 보내지 못함
물론 신뢰성 있는 통신이긴함

