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**Persistence Timer** – timer used to correct 0-size receive window deadlock (see slide 11, SWS)

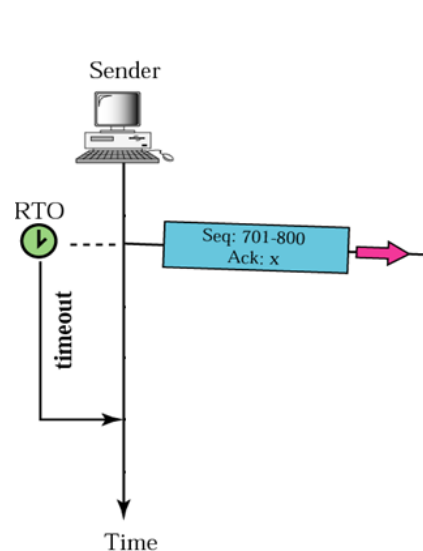
- **assume:**
  - (a) sender receives and ACK with RcvWindow = 0
  - (b) sender stops transmission
  - (c) receiver eventually frees buffer and sends and ACK with non-zero RcvWindow, but **this ACK gets lost!**
- **consequence:** **deadlock** – sender may wait forever!
- **SOLUTION:** each time a sender gets a 0-window segment it starts **persistence timer**
  - (a) if timer goes off send a probe segment with only 1-byte of data
  - (b) probe segments alert receiving TCP that acknowledgment may have got lost and should be resent
- initially **persistence timer = 2\*RTT = retransmission timer**; after that it is doubled until reaching 60 sec; finally one probe every 60 sec ...

## Example [ Persistence Timer ]



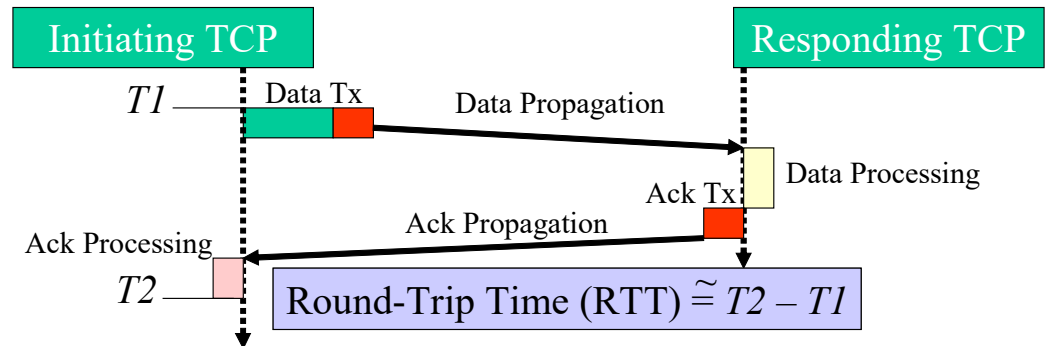
## Retransmission Timer / Timeout (RTO)

- determines how long to wait for ACK of a previously sent segment before retransmission
- depends on distance and network congestion  $\Rightarrow$  **different for each connection, may change during connection-lifetime**



estimated accounts for irregularities in the network

$$\text{RTO} = \text{RTT} + \Delta(\text{RTT}), \text{ or simply } \text{RTO} \approx 2 * \text{RTT}$$
$$\text{RTT} \approx t(\text{ACK received}) - t(\text{data sent})$$



## Calculation of RTT

- TCP determines approximate RTT between devices and adjusts it over time to compensate for increases and decreases
- **does not use a single number – instead, uses dynamic / adaptive formulas that employ a sequence of earlier observations of RTT**

## Average / Smoothed RTT Update Formulas

### 1) Simple Average

$$\overset{\text{estimated}}{\text{ARTT}(K+1)} = \frac{1}{K+1} \sum_{i=1}^{K+1} \overset{\text{observed}}{\text{RTT}(i)}$$

easier to calculate –  
not necessary to  
calculate entire  
summation

$$\text{ARTT}(K+1) = \frac{K}{K+1} \text{ARTT}(K) + \frac{1}{K+1} \text{RTT}(K+1)$$

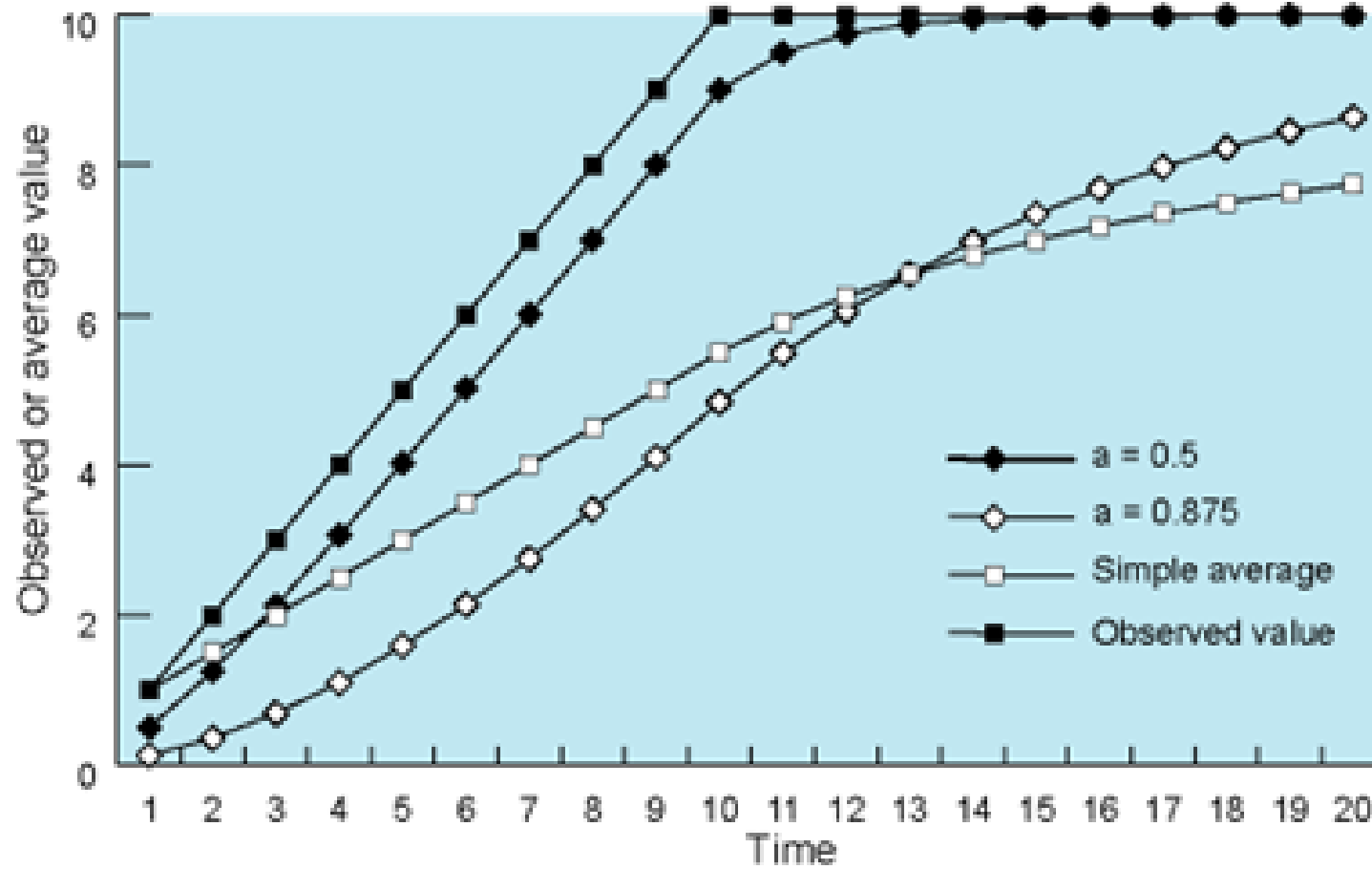
- **drawback:** gives equal weight to each observation
- **solution:** give greater weight to more recent instances since they are more likely to reflect future behavior

### 2) Exponential (Smoothed) Average

$$\text{SRTT}(K+1) = \alpha \cdot \text{SRTT}(K) + (1-\alpha) \cdot \text{RTT}(K+1)$$

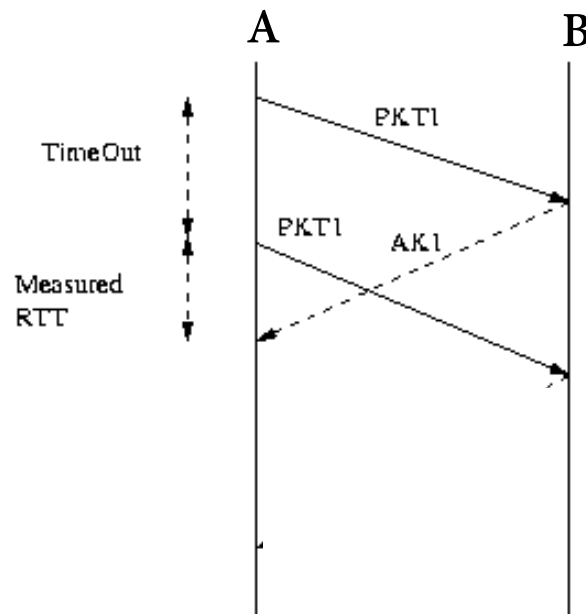
- $0 < \alpha < 1 \Rightarrow$  **smaller  $\alpha$  gives greater weight to more recent observations**
- $\alpha = 0.85 \Rightarrow$  last 10 observations matter
- $\alpha = 0.5 \Rightarrow$  only last 4-5 observations matter

**Example** [ simple vs. exponential average – **significance of choosing right  $\alpha$**  ]



(a) Increasing function

- Karn's Algorithm** – assume an ACK has arrived after retransmission of a TCP segment – does this ACK correspond to the original or retransmitted segment?!
- the source cannot distinguish between two cases
  - misinterpretation can cause RTT to be set much too high or much too low
  - **solution** – Karn's Algorithm: do not update the value of ARTT with RTTs measured for retransmitted segments



# Exercise

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1. In \_\_\_\_\_ data are sent or processed at a very inefficient rate, such as 1 byte at a time.
  - (a) Nagle's syndrome
  - (b) silly window syndrome
  - (c) sliding window syndrome
  - (d) delayed acknowledgment.
  
2. Suppose Host A sends two TCP segments back to back to Host B over a TCP connection. The first segment has a sequence number 90, the second segment has sequence number 110.
  - (a) How much data is in the first segment?
  - (b) Suppose that the first segment is lost but the second segment arrives at B. In the ACK that Host B sends to Host A, what will be the ACK number?