CS414/415 Section 5 Project 4: Reliable networking

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Slides modified from previous years' slides

What do you have to do?

- A connection-based, stream-like reliable communication, similar to TCP.
- Features:
 - Connection-based (open, close etc.).
 - Guarantee that packets are delivered.
 - At least once (not lost in the network).
 - At most once (no duplicates).
 - Guarantees are not absolute (cannot be).

What do you have to do?

- Features (continued):
 - Ordering: deliver in sequence (FIFO).
 - Congestion control (sort of a sliding window with size one).
 - Stream-like (as opposed to packet-like) semantics:
 - Can send data blocks of any size: should fragment.
 - Can receive packets of any size.

How is it related to project #3?

- Unreliable/reliable protocols will co-exist.
 - Two separate APIs, sender can use both simultaneously.
 - Unrel.: "minimsg.h", rel.: "minisocket.h"
 - One may rely on the other, but...
 - ...keep code reasonably separate and isolated!
 - Receiver recognizes type of communication.
 - Demultiplexing: a new protocol type field in the base header, passing control to the right place.

How is it related to project #3?

- Common base abstractions:
 - We are still using ports as communication endpoints for both protocols.
 - Common base packet header (addr., ports)
 - Headers for other protocols staked on top of it
 - We will use miniports for unreliable, sockets for reliable networking.
 - User cannot use same ports for both kinds of communication (miniports hidden in sockets).

A connection-based protocol

- Connection identified by port numbers
- A typical scenario has multiple stages:
 - Server waiting for a client to connect
 - Client connects, information about connection (eg. port numbers) exchanged via hand-shaking
 - Client and server can send data in any direction
 - Client explicitly closes the connection
 - However, server can do it too (eg. when shutting down)
 - After connection closed, no send/receive succeeds
 - Socket is destroyed: accepting more clients requires that a new socket be created

Achieving reliability

- At least once:
 - Delivery confirmation: acknowledgement for every packet received (ACK).
 - Resending if not confirmed (timeout with an exponential back-off).
- At most once:
 - Suppressing duplicates
 - Receiver: duplicate data packets when ACK lost
 - Sender: duplicate ACKs when data believed lost but not really lost (only delayed)
- Not only for data, but also for control packets!
 - Duplicate requests to open/close connection etc.

Ordering and flow control

- Ordering: sequence numbers in packets
 - Normally: buffering, variable-size window, suppressing delivery of packets etc.
- Flow control: dropping packets
 - Normally: we use sliding window with size adjusted to bandwidth
- You can make window size to be equal to one
 - Simplifies implementation tremendously
 - One data packet in transit at all times (very slow)
 - Still need to keep / check sequence numbers

Achieving stream-like semantics

- Send data of any size: fragmentation
 - We simply cut in small pieces (arbitrarily)
 - Assume packet boundaries as determined by the sender app. are meaningless (byte stream)
 - Don't put "reassembling" information in packets
 - Receiver treats all incoming data as parts of a single infinite byte stream
 - Ordering is essential to guarantee correctness here!
 - When data requested, may need to merge data from a few packets to fill the buffer given by client application

Achieving stream-like semantics

- Receive data of any size:
 - Specify maximum amount of data to receive
 - May consume only a part of an incoming packet
 - An "unused" part of the packet is left in the socket for another receive operation to consume
 - May receive less data than requested
 - Since it's a stream, the exact size doesn't matter
 - Client application is assumed smart enough to know what to do with the incoming stream
 - For example, it could add some "merging" information
 - We don't care about it

Achieving stream-like semantics

- One-sender-to-one-receiver, but...
 - ...multiple threads can send, and what's worse...
 - ...multiple threads can issue receive to the same socket concurrently
 - Threads will form a receiver queue to the socket
 - Independent threads can receive random pieces of data!
 - They will need to know how to reassemble them
 - We don't care about it, it's up to the application to assemble
 - Still, all control communication (ACKs etc.) will need to be handled globally, in parallel
 - May require dedicated some threads for this purpose

Minisockets API (to implement)

Creating socket on the server:

- Server installed on a <u>specific port</u> (may fail)
- Waits for incoming connection
 - Blocking (completes only after client connected)
 - Returns a complete socket connected with client
- Simplification: one-to-one communication
 - Only a single client can connect (so it could fail)
 - Once a client connected, connecting not allowed.

Minisockets API (to implement)

void minisocket_destroy(minisocket_t socket);

Our protocol stack

- Base header from "miniports.c" acts as our UDP/IP-like protocol header
 - Need it to identify communication endpoints
 - Need to extend it with protocol type field
- Add a new, TCP-like header on top of that
 - Don't mix info about the reliable communication with the base header!

User application

TCP-like protocol

UDP-like protocol

Network

Implementation approaches

- Approach #1: "TCP" on top of "UDP/IP"
 - "UDP/IP" doesn't know much about "TCP"
 - May need to add miniport "types" to represent various types of protocols stacked on top
 - Still should check if the packets received match the type of receiver (application directly / "TCP")
 - "TCP" uses "UDP" via "send/receive"
 - Any interaction via interface defined in header.
 - May need a separate thread for each port to handle control traffic (sending ACKs etc.)
 - Arguably simpler to implement...
 - ...but that's not the way things are done in real life.

Implementation approaches

- Approach #2: "TCP", "UDP/IP" in parallel
 - Neither of the two protocols "knows" about the other or "uses" the other:
 - Two separate modules for the two protocols
 - Demultiplexing in the network handler:
 - passing control to the right module based on "type"
 - Both are using the same "ports" infrastructure
 - Some code will inevitably be duplicated...
 - ...but we should still keep the two modules reasonably separate, their code shouldn't be intermingled.

Retransmission scheme

- Send, wait for ACK for a given timeout
- Complete if ACK is received on time
 - We don't send ACK to ACK here!
- Resend if ACK not arrived on time
 - Attempt up to 7 times, then give up (error)
 - Start with 100ms delay, then x2 each time
 - If an "old" ACK arrives now, it's still OK

Hand-shaking protocol

- Stages:
 - Client sends OPEN_CONNECTION
 - Server responds with ACK
 - It might also respond with error:
 - Socket in use by another client
 - Socket does not exist
 - Socket exists, not in use, but no thread waiting
 - Client confirms ACK with his own ACK
- Retransmission scheme applies here!

Implementation hints

- Think about it as a state machine
 - States / state:
 - Server: waiting for a client to arrive, client is connected, closing the socket, ...
 - Client: waiting for server to accept connection, ...
 - Server/Client: various stages while sending a packet (ACK received? Which resending attempt? What timeout period?)
 - •
 - State transitions:
 - Packets received
 - An API function called
 - Timeout expired
 - ...

Questions?

- Ask if not sure.
- By all means, come to office hours.