

COMPUTER NETWORKS

LAB ASSIGNMENT-09

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CS22B1095

TCP Server:

```
#include <stdio.h>

#include <stdlib.h>

#include <string.h>

#include <unistd.h>

#include <arpa/inet.h>

#include <time.h>


#define SERVER_PORT 8080

#define MTU_SIZE 1500      // Maximum packet size (MTU)

#define PACKET_LOSS_PROB 0.05  // Probability of packet loss (5%)

#define MAX_LATENCY 100      // Max simulated latency per packet in ms

#define DISPLAY_PACKET_COUNT 20 // Packet count interval for display purposes


// Structure to store each packet's latency data
typedef struct {
    int packet_id;
    int latency;
} PacketLatency;


void receive_file(int client_sock) {
    FILE *file = fopen("received_audio.mp3", "wb");
    if (!file) {
        perror("File open failed");
        close(client_sock);
    }
```

```

    return;
}

char buffer[MTU_SIZE];

int bytes_received;

int packets_received = 0, packets_lost = 0, retransmissions = 0;

int total_latency = 0;

float loss_chance;

srand(time(NULL)); // Seed for random number generator

printf("--- Starting file reception ---\n");

// Receive packets from client
while ((bytes_received = recv(client_sock, buffer, MTU_SIZE, 0)) > 0) {
    packets_received++;

    loss_chance = (float)rand() / RAND_MAX; // Simulate packet loss

    // Simulate packet loss
    if (loss_chance < PACKET_LOSS_PROB) {
        packets_lost++;
        retransmissions++;

        printf("Packet %d lost, simulating retransmission...\n", packets_received);
        continue; // Skip writing this packet to simulate loss
    }

    // Simulate latency
    int latency = rand() % MAX_LATENCY + 1;
    total_latency += latency;
    //usleep(latency * 1000); // Convert latency to microseconds

    fwrite(buffer, sizeof(char), bytes_received, file);
}

```

```

// Optional: Display every N packets received
if (packets_received % DISPLAY_PACKET_COUNT == 0) {
    printf("Received %d packets so far...\n", packets_received);
}
}

if (bytes_received < 0) {
    perror("Error receiving data");
} else {
    printf("File received successfully.\n");
}

// Calculate and print summary statistics
float average_latency = packets_received > 0 ? (float)total_latency / packets_received : 0;
printf("\n--- Transmission Summary ---\n");
printf("Total Packets Received: %d\n", packets_received);
printf("Total Packets Lost: %d\n", packets_lost);
printf("Total Retransmissions: %d\n", retransmissions);
printf("Average Latency per Packet: %.2f ms\n", average_latency);
printf("Total Transmission Latency: %d ms\n", total_latency);
printf("Efficiency : %.2f%%\n", (float)(packets_received - packets_lost) / packets_received * 100);
printf("Packet Loss Rate: %.2f%%\n", (float)packets_lost / packets_received * 100);
printf("Reassemble reliability: %.2f%%\n", (float)(packets_received -
packets_lost) / packets_received * 100);
fclose(file);
}

int main() {
    int server_sock, client_sock;
    struct sockaddr_in server_addr, client_addr;

```

```

socklen_t addr_size = sizeof(client_addr);

// Create TCP socket
server_sock = socket(AF_INET, SOCK_STREAM, 0);
if (server_sock < 0) {
    perror("Socket creation failed");
    return 1;
}

// Set up server address structure
server_addr.sin_family = AF_INET;
server_addr.sin_port = htons(SERVER_PORT);
server_addr.sin_addr.s_addr = INADDR_ANY;

// Bind the socket to the specified IP and port
if (bind(server_sock, (struct sockaddr *)&server_addr, sizeof(server_addr)) < 0) {
    perror("Bind failed");
    close(server_sock);
    return 1;
}

// Listen for incoming connections
if (listen(server_sock, 1) < 0) {
    perror("Listen failed");
    close(server_sock);
    return 1;
}

printf("Server listening on port %d...\n", SERVER_PORT);

// Accept a client connection
client_sock = accept(server_sock, (struct sockaddr *)&client_addr, &addr_size);

```

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    if (client_sock < 0) {
        perror("Accept failed");
        close(server_sock);
        return 1;
    }
    printf("Client connected.\n");

    // Start receiving file from the client
    receive_file(client_sock);

    // Close sockets
    close(client_sock);
    close(server_sock);
    return 0;
}

```

TCP Client:

```

#include <stdio.h>
#include <stdlib.h>
#include <string.h>
#include <unistd.h>
#include <arpa/inet.h>
#include <errno.h>

#define SERVER_IP "127.0.0.1" // Change to server's IP if not running locally
#define SERVER_PORT 8080
#define MTU_SIZE 1500 // Maximum Transmission Unit (packet size)

void send_file(int server_sock, const char *filename) {

```

```

FILE *file = fopen(filename, "rb");

if (!file) {
    perror("File open failed");
    close(server_sock);
    return;
}

char buffer[MTU_SIZE];

int bytes_read;
int total_packets = 0;

printf("Starting file transmission...\n");

// Read from file and send data in chunks
while ((bytes_read = fread(buffer, sizeof(char), MTU_SIZE, file)) > 0) {
    if (send(server_sock, buffer, bytes_read, 0) < 0) {
        perror("Failed to send packet");
        fclose(file);
        close(server_sock);
        return;
    }
    total_packets++;

    // Optional: Display progress every few packets
    if (total_packets % 20 == 0) {
        printf("Sent %d packets so far...\n", total_packets);
    }
}

printf("File transmission completed. Total packets sent: %d\n", total_packets);

```

```

    fclose(file);
}

int main() {
    int client_sock;
    struct sockaddr_in server_addr;

    // Create TCP socket
    client_sock = socket(AF_INET, SOCK_STREAM, 0);
    if (client_sock < 0) {
        perror("Socket creation failed");
        return 1;
    }

    // Set up server address structure
    server_addr.sin_family = AF_INET;
    server_addr.sin_port = htons(SERVER_PORT);
    if (inet_pton(AF_INET, SERVER_IP, &server_addr.sin_addr) <= 0) {
        perror("Invalid address/ Address not supported");
        close(client_sock);
        return 1;
    }

    // Connect to the server
    if (connect(client_sock, (struct sockaddr *)&server_addr, sizeof(server_addr)) < 0) {
        perror("Connection to server failed");
        close(client_sock);
        return 1;
    }

    printf("Connected to server %s on port %d.\n", SERVER_IP, SERVER_PORT);
}

```

```

// Send file to the server

send_file(client_sock, "birds-chirping-75156.mp3"); // Replace with your audio file path


close(client_sock);

return 0;

}

```

```

Packet 200 lost, simulating retransmission...
Received 220 packets so far...
Packet 229 lost, simulating retransmission...
Received 240 packets so far...
Packet 253 lost, simulating retransmission...
Packet 257 lost, simulating retransmission...
Received 260 packets so far...
Packet 276 lost, simulating retransmission...
Packet 279 lost, simulating retransmission...
Received 280 packets so far...
Received 300 packets so far...
File received successfully.

```

```

--- Transmission Summary ---
Total Packets Received: 300
Total Packets Lost: 13
Total Retransmissions: 13
Average Latency per Packet: 47.63 ms
Total Transmission Latency: 14289 ms
Efficiency : 95.67%
Packet Loss Rate: 4.33%
Reassemble reliability: 95.67%

```

```

Sent 300 packets so far...
File transmission completed. Total packets sent: 300
charish@LAPTOP-GFCS9LJ9:~/cn/LAB 09$ gcc audio_tcp_client.c -o b
charish@LAPTOP-GFCS9LJ9:~/cn/LAB 09$ ./b
Connected to server 127.0.0.1 on port 8080.
Starting file transmission...
Sent 20 packets so far...
Sent 40 packets so far...
Sent 60 packets so far...
Sent 80 packets so far...
Sent 100 packets so far...
Sent 120 packets so far...
Sent 140 packets so far...
Sent 160 packets so far...
Sent 180 packets so far...
Sent 200 packets so far...
Sent 220 packets so far...
Sent 240 packets so far...
Sent 260 packets so far...
Sent 280 packets so far...
Sent 300 packets so far...
File transmission completed. Total packets sent: 300

```


UDP Server:

```
#include <stdio.h>

#include <stdlib.h>

#include <string.h>

#include <unistd.h>

#include <sys/time.h>

#include <arpa/inet.h>

#include <errno.h>

#include <time.h>


#define SERVER_PORT 8081

#define MTU_SIZE 1500

#define PACKET_LOSS_PROB 0.05


typedef struct {
    int packet_id;
    int latency;
} PacketLatency;


void receive_file(int udp_sock, int total_packets) {
    FILE *file = fopen("received_audio_udp.mp3", "wb");
    if (!file) {
        perror("File open failed");
        close(udp_sock);
        return;
    }

    struct sockaddr_in client_addr;
```

```

socklen_t addr_size = sizeof(client_addr);

char buffer[MTU_SIZE];

int bytes_received;

int packets_received = 0, packets_lost = 0, retransmissions = 0;

int total_latency = 0;


srand(time(NULL));


printf("Waiting for UDP packets...\n");


// Set socket timeout
struct timeval timeout;

timeout.tv_sec = 5; // Wait for 5 seconds of inactivity before timing out
timeout.tv_usec = 0;

setsockopt(udp_sock, SOL_SOCKET, SO_RCVTIMEO, &timeout, sizeof(timeout));


while ((bytes_received = recvfrom(udp_sock, buffer, MTU_SIZE, 0, (struct sockaddr *)&client_addr,
&addr_size)) > 0) {
    packets_received++;

    printf("Received packet %d with %d bytes\n", packets_received, bytes_received);


    float loss_chance = (float)rand() / RAND_MAX;

    if (loss_chance < PACKET_LOSS_PROB) {
        printf("Simulated packet loss for packet %d\n", packets_received);
        packets_lost++;
        retransmissions++;
        continue;
    }


    int latency = rand() % 100 + 1;

```

```

    total_latency += latency;

    fwrite(buffer, sizeof(char), bytes_received, file);
}

if (bytes_received < 0 && errno == EWOULDBLOCK) {
    printf("Timeout reached. No more packets are being received.\n");
}

// Calculate average latency
float average_latency = packets_received > 0 ? (float)total_latency / packets_received : 0;

// Print the summary
printf("\n--- Transmission Summary ---\n");
printf("Total Packets Received: %d\n", packets_received);
printf("Total Packets Lost: %d\n", packets_lost);
printf("Total Retransmissions: %d\n", retransmissions);
printf("Average Latency per Packet: %.2f ms\n", average_latency);
printf("Total Transmission Latency: %d ms\n", total_latency);
printf("Efficiency : %.2f%%\n", (float)(packets_received-packets_lost)/packets_received*100);
printf("Packet Loss Rate: %.2f%%\n", (float)packets_lost / packets_received * 100);
printf("Reassemble reliability: %.2f%%\n", (float)(packets_received-
packets_lost)/packets_received*100);

fclose(file);
}

int main() {
    int udp_sock;
    struct sockaddr_in server_addr;

```

```

udp_sock = socket(AF_INET, SOCK_DGRAM, 0);
if (udp_sock < 0) {
    perror("Socket creation failed");
    return 1;
}

server_addr.sin_family = AF_INET;
server_addr.sin_port = htons(SERVER_PORT);
server_addr.sin_addr.s_addr = INADDR_ANY;

if (bind(udp_sock, (struct sockaddr *)&server_addr, sizeof(server_addr)) < 0) {
    perror("Bind failed");
    close(udp_sock);
    return 1;
}

printf("UDP Server listening on port %d...\n", SERVER_PORT);

int total_packets = 1000;
receive_file(udp_sock, total_packets);

close(udp_sock);
return 0;
}

```

UDP Client:

```

#include <stdio.h>
#include <stdlib.h>
#include <string.h>
#include <unistd.h>
#include <arpa/inet.h>

```

```

#include <errno.h>

#define SERVER_IP "127.0.0.1" // Change to the server's IP if not running locally
#define SERVER_PORT 8081
#define MTU_SIZE 1500 // Maximum Transmission Unit (packet size)

void send_file(int udp_sock, const char *filename, struct sockaddr_in server_addr) {
    FILE *file = fopen(filename, "rb");

    if (!file) {
        perror("File open failed");
        close(udp_sock);
        return;
    }

    char buffer[MTU_SIZE];
    int bytes_read;
    int packet_id = 0;

    printf("Starting file transmission...\n");

    // Read from file and send data in packets
    while ((bytes_read = fread(buffer, sizeof(char), MTU_SIZE, file)) > 0) {
        // Add packet ID to buffer for identification (optional)
        memcpy(buffer, &packet_id, sizeof(packet_id));

        if (sendto(udp_sock, buffer, bytes_read + sizeof(packet_id), 0, (struct sockaddr *)&server_addr,
        sizeof(server_addr)) < 0) {
            perror("Failed to send packet");
            fclose(file);
            close(udp_sock);
            return;
        }
    }
}

```

```

    }

    packet_id++;

    // Optional: Display progress every few packets
    if (packet_id % 20 == 0) {
        printf("Sent %d packets so far...\n", packet_id);
    }
}

printf("File transmission completed. Total packets sent: %d\n", packet_id);

fclose(file);
}

int main() {
    int udp_sock;
    struct sockaddr_in server_addr;

    // Create UDP socket
    udp_sock = socket(AF_INET, SOCK_DGRAM, 0);
    if (udp_sock < 0) {
        perror("Socket creation failed");
        return 1;
    }

    // Set up server address structure
    server_addr.sin_family = AF_INET;
    server_addr.sin_port = htons(SERVER_PORT);
    if (inet_pton(AF_INET, SERVER_IP, &server_addr.sin_addr) <= 0) {
        perror("Invalid address/ Address not supported");
    }
}

```

```

        close(udp_sock);

        return 1;
    }

    printf("Sending file to UDP server at %s:%d\n", SERVER_IP, SERVER_PORT);

    // Send file to the server

    send_file(udp_sock, "birds-chirping-75156.mp3", server_addr); // Replace with your audio file
    path

    close(udp_sock);

    return 0;
}

```

```

Received packet 276 with 1500 bytes
Received packet 277 with 1500 bytes
Received packet 278 with 1500 bytes
Received packet 279 with 1500 bytes
Received packet 280 with 1500 bytes
Received packet 281 with 1500 bytes
Received packet 282 with 1500 bytes
Received packet 283 with 1500 bytes
Received packet 284 with 1500 bytes
Received packet 285 with 1500 bytes
Received packet 286 with 1500 bytes
Received packet 287 with 1500 bytes
Received packet 288 with 1500 bytes
Received packet 289 with 1500 bytes
Received packet 290 with 1500 bytes
Received packet 291 with 1500 bytes
Received packet 292 with 1500 bytes
Received packet 293 with 1500 bytes
Received packet 294 with 1500 bytes
Received packet 295 with 1500 bytes
Received packet 296 with 1500 bytes
Received packet 297 with 1500 bytes
Received packet 298 with 1500 bytes
Received packet 299 with 1500 bytes
Received packet 300 with 304 bytes
Timeout reached. No more packets are being received.

```

```

--- Transmission Summary ---
Total Packets Received: 300
Total Packets Lost: 15
Total Retransmissions: 15
Average Latency per Packet: 48.03 ms
Total Transmission Latency: 14410 ms
Efficiency : 95.00%
Packet Loss Rate: 5.00%
Reassemble reliability: 95.00%

```

```

charish@LAPTOP-GFCS9LJ9:~/cn/LAB 09$ gcc audio_udp_client.c -o b
charish@LAPTOP-GFCS9LJ9:~/cn/LAB 09$ ./b
Sending file to UDP server at 127.0.0.1:8081
Starting file transmission...
Sent 20 packets so far...
Sent 40 packets so far...
Sent 60 packets so far...
Sent 80 packets so far...
Sent 100 packets so far...
Sent 120 packets so far...
Sent 140 packets so far...
Sent 160 packets so far...
Sent 180 packets so far...
Sent 200 packets so far...
Sent 220 packets so far...
Sent 240 packets so far...
Sent 260 packets so far...
Sent 280 packets so far...
Sent 300 packets so far...
File transmission completed. Total packets sent: 300
charish@LAPTOP-GFCS9LJ9:~/cn/LAB 09$

```

Network fragmentation:

Smaller MTU sizes lead to more packets and thus increased fragmentation, which could impact transmission performance and reassembly.

Impact on TCP and UDP:

TCP - handles packet reassembly with error correction.

- Smaller MTU sizes may lead to more frequent acknowledgments and re-transmissions in the case of packet loss, affecting overall throughput.

UDP - does not guarantee packet delivery or reassembly.

- Smaller MTU sizes mean more packets are likely to be lost, especially if network conditions are suboptimal, leading to reduced reliability in UDP-based transmissions.

Larger MTUs result in fewer packets, which can be more efficient but may encounter issues if a packet is too large to be sent in one piece, especially in networks with lower MTU limits.

Smaller packets might reduce the impact of any single packet loss in TCP but could lead to delays in UDP where lost packets aren't re-transmitted. TCP's reassembly mechanism ensures that lost packets are recovered, while UDP's lack of such guarantees may lead to data loss in audio streams.

Packet Loss: Smaller MTUs mean more packets, increasing the risk of individual packet loss but potentially reducing the impact on overall data, as each packet carries less.

Latency: More packets can introduce latency, especially with TCP's need for acknowledgment.

Re-Transmissions: TCP will attempt to resend lost packets, while UDP will not. With smaller MTUs, TCP re-transmissions may occur more frequently.

Real-Time Audio: UDP is preferred due to lower latency, but small MTU sizes could lead to packet loss, which impacts audio quality. Balancing MTU size is critical for minimizing latency while ensuring data is transmitted reliably.

Non-Real-Time Audio: TCP may be better suited for ensuring complete data delivery, even with a higher MTU size, as latency is less critical.