CIS 505: Software Systems

Spring 2017

Assignment 3: Chat server

Due on March 29th, 2017, at 10:00pm EDT

1 Overview

For this assignment, you will implement a simple replicated chat server that uses multicast to distribute the chat messages to the different replicas.

The figure on the right shows a sketch of the system you will build. There are two kinds of processes: servers (represented by orange squares) and clients (represented by yellow circles). Each client is connected to *one* of the servers, and each server is connected to *all* the other servers. The clients and servers can all run on different machines, or (e.g., for testing) they can all run on the same machine, e.g., in your VM.

The system will provide different chat rooms. Client can join one of the chat rooms and then post messages, which should be delivered to all the other clients that are currently in the same chat room. If there was only a single server, this would be easy: the server would simply send each message it receives to all the clients who are currently in the relevant room. However, since there can be multiple servers, this simple approach does not work. Instead, the servers will internally use multicast to deliver the messages.

To make this a little more concrete, consider the following example. In the figure, client D is connected to server S₃. Suppose D joins chat room #3, which already contains A and B, and posts a message there (say, "Hello"). S₃ then multicasts D's message to multicast group #3, so it will be delivered to the other servers. (All servers are members of all multicast groups.) The servers then forward D's message to all of the clients that are currently in chat room #3. In this case, S₃ will echo it back to D, and S₁ will send it to A and B. S₂ does not currently have any clients that are in chat room #3, so it ignores the message.

The servers should support three different ordering modes that we studied in class:

- Unordered multicast: Each server delivers the messages to its clients in whatever order the server happens to receive it;
- **FIFO multicast:** The messages from a given client C should be delivered to the other clients in the order they were sent by C; and
- **Totally ordered multicast:** The messages in a given chat room should be delivered to all the clients in that room in the same order.

Your implementation should be fully distributed; in other words, there should not be any central 'sequencing nodes'. All communication, both between clients and servers and among the servers, should be via *unicast* UDP (and not using the OS's built-in multicast primitive). You may assume that there is no message loss, and that the messages between servers and clients are not reordered.

2 Specification

2.1 The client

Your first task is to implement a simple chat client. The user should be able to invoke the client from the command line with the IP address and port of a particular server as a parameter. (Each client is connected to exactly one server.) The user should then be able to type lines of text (commands or chat messages), and, whenever the user has entered a new line, the chat client should send the line to the server in a single UDP packet (without any trailing CR or LF). The client should also listen for UDP packets from the server; whenever it receives such a packet, it should print its contents to the terminal.

2.2 Commands

When the server receives a line of text from the user, it should check whether it begins with a forward slash (/). If not, the server should consider the line to be a chat message; otherwise, it should treat it as a command. The following commands should be supported:

- /join <roomID>: Used to join a particular chat room (Example: /join 7). The roomID should be an integer in the set {1, 2, ..., N}, where N is the total number of chat rooms. Your implementation should support at least N = 10 chat rooms (i.e., with room IDs 1,2,3,...,10). If the command succeeds, the server should respond with a line that starts with +OK (Example: +OK You are now in chat room #7). If the user tries to join more than one chat room at a time, or if the specified chat room does not exist (e.g., the specified room is outside your supported range of room IDs), the server should respond with a line that starts with -ERR (Example: -ERR You are already in room #9).
- /part: Used the leave the current chat room. If successful, the server sends a response that starts with +OK (Example: +OK You have left chat room #7). If user is not currently in a chat room, the server sends a response that starts with -ERR.
- /nick: Used to set the user's nickname (Example: /nick Linh). The server should respond with a line that starts with +OK. If the user does not explicitly set a nickname, the server should use its IP and port (Example: 127.0.0.1:5005). If the /nick command is used more than once, the server should use the latest nickname.
- /quit: When the user enters this special command, the chat client should send the command to the server and then terminate. The server should remove the chat client from its list of active clients.

If the user sends a text message without having joined a chat room, the server should respond with a line that starts with -ERR. To make your implementation easier, each client cannot be in more than one chat room at a time.

Below is an example session (the lines in bold are typed by the user):

```
$ ./chatclient 127.0.0.1:8000
/join 3
+OK You are now in chat room #3
```

```
Hello everyone!
<127.0.0.1:57326> Hello everyone!
<Alice> Hi there! Do you want to set a nickname?
/nick Bob
+OK Nickname set to 'Bob'
Is this better?
<Bob> Is this better?
<Alice> Yes, much better!
/join 5
-ERR You are already in room #3
/part
+OK You have left chat room #3
/quit
```

2.3 The server

Often, multicast systems support dynamic membership: nodes can join and leave the system at runtime. However, to simplify the assignment a bit, we will implement only static membership: each server will have access to a configuration file that contains the addresses of all the other servers.

Each server will have two – potentially different – addresses. The first is the *forwarding address*: this is the address that other servers should use as the destination address when they want to send a message to this server. The second is the *bind address*: this is the IP and port number on which the server listens for UDP messages from clients and from other servers. Usually, the two are the same. However, for debugging, it is useful to have the ability to make them different; for more details, see Section 3.2.

Your server program should require two command-line arguments. The first is the name of a configuration file that contains the forwarding address and the bind address of *all* the server instances, and the second is the index of the current server instance in this list. The file should contain one line for each server instance. If the line contains only one IP address and port number (Example: 127.0.0.1:8000), this address should serve as both the forwarding address and the bind address for that server. If the line contains two IPs and port numbers separated by a comma (Example: 127.0.0.1:8001, 127.0.0.1:5001), the first address is the forwarding address and the second is the bind address. The index of the first server should be 1. For instance, suppose the file foo.txt contains the following lines:

```
127.0.0.1:8000,127.0.0.1:5000
127.0.0.1:8001,127.0.0.1:5001
127.0.0.1:8002,127.0.0.1:5002
```

and the server is invoked like this: chatserver foo.txt 2, then the server would listen for messages on (its bind address) 127.0.0.1:5001. When it has a message for the third server, it would send the message to (the third server's forwarding address) 127.0.0.1:8002, and when it has a message for the first server, it sends it to 127.0.0.1:8000.

The server should also accept two options (-v and -o) on the command line. When the -v option is present, the server should print a line to stdout when something important happens. (This is meant for debugging, so it is up to you to decide what qualifies as important.) Each such line should start with a prefix of the form hh:mm:ss.uuuuuu Snn, where hh:mm:ss.uuuuuu is the current time in hours, minutes, seconds, and microseconds, and nn is the server index from the command line. For instance, the

server might print 03:48:22.004328 S02 Client 17 posts "Hi" to chat room #3. The -o option is used to select an ordering mode: choices are -o unordered (the default), -o fifo, and -o total. When no arguments are given on the command line at all, the server should print your name and SEAS login and then terminate.

When a client posts a message to a chat room, the server should prepend " $< \times \times \times >$ " to the message, where $\times \times \times$ is the client's nickname, just like in the example transcript above. Your implementation should be able to support at least 10 servers, with up to 250 clients total.

3 Suggestions

The following section contains some hints that we thought you might find helpful as you implement, test, and debug your solution. None of this is mandatory – feel free to use your own approach if you prefer!

3.1 Implementation hints

The client is a good place to begin. The slide deck about socket programming already contains some code for a simple UDP client, so you could start with that. However, notice that the chat client needs to listen not just for messages from the server but *also* for user input. One simple way to implement this is with the select system call: you can use this to block until *either* the UDP socket *or* stdin are ready for reading. The file descriptor number for stdin is STDIN_FILENO; once this descriptor becomes ready for reading, you can use read from it as usual, and you should receive a line of user input.

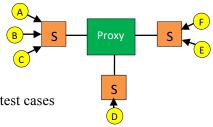
You can initially test your client with a simple UDP echo server, similar to the one on the slides. Once the echo is working, you can add code for reading the file with the server addresses, and forward incoming client messages to the other servers, who can forward them to their own clients. (Remember: This should be done using unicast UDP, not using the kernel's built-in multicast primitive!) The result is, effectively, a simple unordered multicast with only one group: a message sent by any client will be forwarded to all the clients. Once this is working, a good next step would be to add support for /join, /part, etc. Notice that, once you have support for groups, the servers cannot simply forward the messages they receive; they need to attach some additional information, such as the group to which the message belongs. We recommend that you use a text format for your messages: for instance, if one server wants to tell another about a Hello message in chat room #5, it could send something like 5, Hello. This makes it easy to print messages into the log (with the -v option) and to inspect them when something goes wrong.

At this point, the big missing pieces are FIFO and total ordering. These will require a holdback queue, as we have discussed in detail in class. Start with FIFO ordering; this will probably require extending your inter-server messages with even more information. Notice that it isn't completely trivial to test whether the ordering is working properly, so you may want to use the proxy for testing (see Section 3.2).

The final piece of the puzzle is total ordering. This will require some more complicated messaging between the servers; the algorithm is on the slides of Lecture 11. Don't forget to include a proper tie breaker! The IP address isn't enough for this: for instance, if you test your server in the VM, all instances will have the same IP address. For instance, a combination of IP and port number should work.

3.2 Testing your solution

Testing a system like this is a little tricky. You'll probably want to create a few servers and clients within your VM (they could each run in a separate terminal window, for instance). However, within the VM, network messages are delivered almost instantaneously, so, by the time you enter the next message, the previous one will already have been delivered to all the clients. Thus, it is hard to produce 'interesting' test cases where several messages are in flight at the same time.



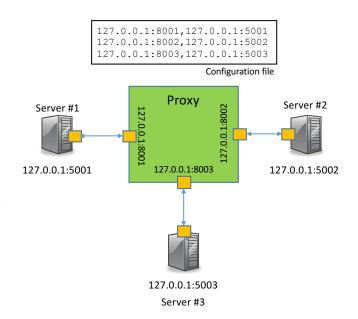
To help you with testing, we have provided two tools in the test/ subdirectory. The first is called proxy.cc. You can insert the proxy between the servers, as shown in the illustration above; it will receive all the messages that the servers send to each other, and insert a random delay, which can be configured (in microseconds) with the -d option. If you make the delay large enough (say, a few seconds), the messages should frequently arrive out of order, which allows you to properly test all the interesting cases. (Please note that the code for the proxy is already provided; all you need to do is to invoke it together with your servers for testing.)

For the proxy to work, the forwarding address and the bind address have to be different: the servers themselves will listen on the bind addresses, while the proxy listens on the forwarding addresses and then forwards any messages it receives to the bind address of the destination server. When you use the proxy, create an address file that contains two different addresses on each line (it can be 127.0.0.1 with different port numbers) and give the same file to both the servers and the proxy.

An illustration of using the proxy:

The figure on the right shows a simple configuration file with three servers, as well as the communication pattern that should be used in this case. Notice how each server only has one socket, which is bound to the bind address, and how the proxy has a socket for each of the servers, which is bound to the forwarding address.

Suppose server #1 wants to send a message to server #2. Then the following should happen: Server #1 sends the message to server #2's forwarding address. The resulting UDP packet has source address 127.0.0.1:5001 and destination address 127.0.0.1:8002, since it is going to server



#2. (Don't let the arrows in the picture confuse you - each server can send messages to each of the proxy's ports!). The message is received by the proxy on port 127.0.0.1:8002. Since this is the forwarding address of server #2, the proxy knows that this message was supposed to go to that server. Also, since the source address is server #1's bind address, the proxy knows that the message came from server #1. The

proxy forwards the message to server #2, using the socket that is bound to server #1's forwarding address (127.0.0.1:8001). The resulting UDP packet has source address 127.0.0.1:8001 and destination address 127.0.0.1:5002. (Again, don't let the arrows in the picture confuse you!). The message is received by server #2. Since its source address is server #1's forwarding address, server #2 knows that this is where the message (originally) came from.

3.3 Testing your solution with the tester

Another challenge with testing comes from the fact that it is tedious to manually send lots of test messages, and that it can be difficult to see whether a long string of messages was ordered correctly. This is where the second tool, stresstest.cc (provided in the test/ subdirectory) comes in.

At a high level, the stress tester is a single program that can emulate multiple clients. For instance, when you run it like this:

```
./stresstest -o fifo -m 20 -c 5 -q 2 servers.txt
```

it will emulate five clients (-c) that each join one of two (-g) chat rooms and then post a total of 20 messages (-m). It will read the list of servers from 'servers.txt'; the format is the same as in the configuration file you give to the server processes. The stress tester will examine the messages it receives from your servers, and it can check whether they are ordered correctly; you can specify an order with the -o option, which works just like the -o option on the servers. (When you use -o unordered, the stress tester only checks whether each client receives all of the messages that are being posted.) If something goes wrong, you may want to re-run the stress tester with the -v option, which produces a much more verbose output. The tester should be used in conjunction with the proxy. (As explained above, without the proxy, the communication delay will be so low that you rarely get message reordering; as a result, the messages will naturally appear to be ordered, even if your implementation does not work correctly.) Please make sure you have at least tested your solution with the stress tester before submitting.

3.4 Debugging hints

Debugging truly distributed systems, such as this one, can be tricky because many things are happening concurrently on different nodes. The -v option is intended to help you with this. If you encounter a problem (say, some messages are not being delivered, or the proposed sequence numbers look wrong), you can use -v on all the servers and redirect the output of each server to a file. Then you can cat the files together and sort them. Because each line starts with a timestamp and a node ID, this will result in a kind of chronicle that shows all the events on the different servers in the order in which they occurred.

For this to work well, it is important to produce sufficiently many (and sufficiently detailed) log messages. Also, it is usually a good idea to include periodic 'dumps' of key data structures, such as the holdback queue, so you can see the status of the various messages at different points in time.

4 Submitting your solution

Before you submit your solution, please make sure that:

Your solution compiles properly.
Your code contains a reasonable amount of useful documentation (required for style points).
You have completed <i>all</i> the fields in the README file.
You have checked your final code into the Git repository before you submit.
You are submitting a . zip file (e.g., one created by running make pack from the HW3 folder)
Your .zip file contains all of the following:
□ all the files needed to compile your solution (especially all .cc files!);
□ a working Makefile; and
☐ the README file, with all fields completed
Your . zip file is smaller than 250kB. Please do not submit binaries or input files!

As with the previous assignments, you must submit your .zip file online, using the web submission system - we will not accept submissions via email, or in any other way. If you need an extension and still have jokers available, you can use the submission system to extend the deadline.

5 Extra credit

5.1 Causal ordering (+15%)

For this extra-credit task, you should add a fourth ordering mode: causal ordering. As with the existing three modes, your implementation must be fully distributed – for instance, you may not use a central coordinator. Your server should support an -o causal option to enable this mode.

5.2 Lossy links (+5%)

For this extra-credit task, you should add a way to handle message losses on the client-server links as well as on the server-server links. Lost messages should be retransmitted, but you may assume that the network remains connected, and that a message will eventually be delivered if it is retransmitted sufficiently often. The proxy.cc tool already supports a command-line option -1 that causes the proxy to drop a certain fraction of the messages it forwards; this should be useful for testing.

6 Grading

The maximum score for the main part of the assignment is 100 points. It will be distributed as follows:

10%: Style (README, printing the name, etc.) and documentation

10%: Chat client functionality

40%: Basic server functionality

40%: Ordering protocols

The maximum number of extra-credit points is 20, if you complete the tasks in both 5.1 and 5.2.