

ClouDJ: A Music Sharing Application

Amanda Hittson, Rebecca Lam, Josh Slauson
University of Wisconsin-Madison

ABSTRACT

ClouDJ enables music sharing. Listening to music is largely a social activity that is best enjoyed with others. However, there is currently no easy way to listen to music together online. We plan to make listening and sharing music easier with our system ClouDJ. ClouDJ will allow users to play their own music while sharing it with other users in real-time.

1. INTRODUCTION

For our project we designed and implemented ClouDJ, a music sharing application that uses a centralized server system. The ClouDJ application allows multiple clients to simultaneously listen to music together in shared sessions. To this end, we designed a system with the following goals:

- **Performance:** users that are participating in a session together should be able to listen to music at a similar rate.
- **Availability:** users must be able to continuously listen to music in a session assuming the user hosting the session is available.
- **Scalability:** adding clients should not affect system performance.

The rest of the paper is structured as follows. In Section 2 we discuss the architecture and design of the underlying components. In Section 3 we examine the implementation details and how we adapt the design to the Google App Engine framework. In Section 4 we evaluate our system based on the aforementioned performance goals, and in Section 5 and Section 6 we finish with future work and concluding remarks.

2. DESIGN

To create ClouDJ, we implemented the design described in subsections 2.1 and 2.2 by integrating Google App Engine. Figure 1 depicts the overall system.

2.1 Architecture

There are three major roles in our system: the master handler, session handler, and client. Clients own and store music, can share their music with others and can listen to music shared by others. The master handler provides access control, starts sessions for users who want to host, and acts as a liaison between the clients and session servers. The session handler is in charge of handling synchronization messages between clients and serving content.

2.1.1 Front End

A session is an abstraction in which multiple users may listen to one song that is hosted by a single user. Our front end (client) application allows a user to either create a session and become a host, or join an existing session and become a listener. When acting as a host, a user may add songs from the session playlist, play or pause the current song, and skip to the next song. When acting as a listener, the user has no control over what song is being played. Users can see all sessions that they are able to join. Only members of a user's access control list (ACL) may see or join any session in which that user is a host.

The client has access to its user's music and playlists and also keeps track of data such as the user's current session and the user's potential sessions (sessions this user can access). For the current session, the client keeps track of the current session key and relevant song information. For the potential sessions, the client keeps the host, session key, and currently playing song.

2.1.2 Back End

The backend infrastructure is more complicated than the frontend. The master server keeps track of users currently online, user ACLs, and user membership lists (ACLs it is a member of). It also is responsible for maintaining a session table, a table that maps host users to sessions (this is a one-to-one mapping). When a client logs on, the master server informs each session associated to a host on the client's membership list that this client is a potential listener.

The session handler is the workhorse of the system. It services requests for sessions it is in charge of by routing data from the host client to relevant clients (listeners). It maintains the list of listeners and potential listeners. A potential listener is a client who could listen to this session, but is currently not. In other words, these are the clients on the host client's ACL that are logged in but not listening to this session. Session servers also take care of session cleanup when a session ends (or fails).

2.1.3 Storage

Certain information is stored on the server in the Datastore and Blobstore. Datastore holds session data and ACLs. Session data includes session meta data (session key, host, listeners), and session state (current song, song play or pause flag, session ended flag, timestamp of last play/next message). This information is used to coordinate sessions among different clients. The ACLs keeps track of the potential sessions and potential listeners for each user. Potential listen-

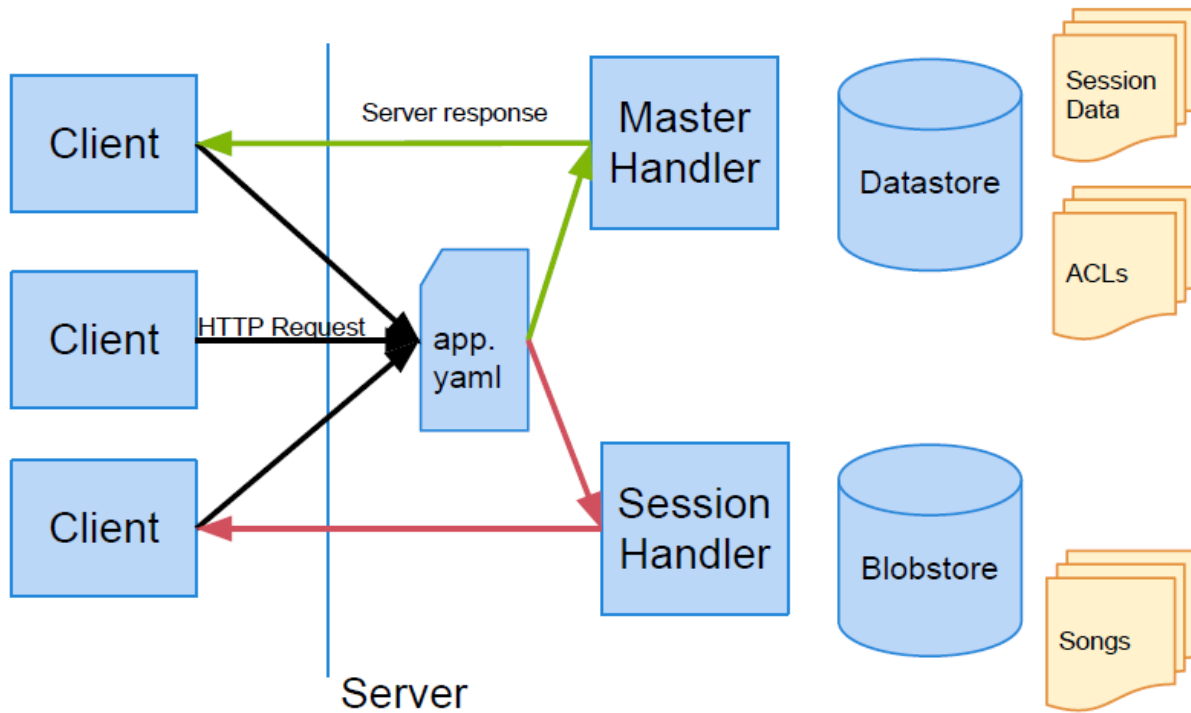


Figure 1: The system architecture.

ers are users who can listen to a session hosted by a given user and potential sessions are sessions the user can join. Blobstore is only used to store song data in our application. When a session ends, the songs are deleted from the blobstore.

2.2 Execution Flow

As shown in Figure 1, the general execution of the ClouDJ application is as follows:

1. The client contacts the server
2. The server performs executes the handler based on the request
3. The handler runs and propagates updates to session participants
4. The client receives the server response and performs actions

In sections 2.2.1 and 2.2.2, we discuss session creation, joining and leaving sessions, and updating sessions.

2.2.1 Creating, Joining, and Leaving a Session

Sessions are created when a client contacts the master handler about hosting a new session. The master handler then creates a new session in the datastore and responds to the client, which updates its current session information. Clients may join sessions by contacting the master handler with the session key of the session they wish to join. The master handler updates the session information in the datastore and notifies everyone in the session of the new listener. After a session is established, the client no longer communicates with the master handler. When a user leaves a session,

the client contacts the session handler telling it to remove itself from the listener list if it is not the host. If it is the host the session handler notifies all listeners that the session has ended and removes the session and corresponding song data from the datastore and blobstore.

2.2.2 Updating Sessions

Hosts may update the sessions using several types of messages: “upload”, “play”, “pause,” and “next,”. Songs are added to the session when the host client contacts the session handler with an “upload” message, indicating that it wants to add a song to the playlist. The session handler forwards the song to the blobstore and retrieves the corresponding blob key, which is used when requesting the song for playback. The session handler then updates the session playlist in the datastore and sends the blob key to all clients in the session. The clients then fetch the song from the blobstore.

Not surprisingly, the “play”, “pause,” and “next” messages indicate when the host wants to unpause, pause, and skip the current song, respectively. The host sends these control messages to the session handler, which updates the session in the datastore and then propagates the message to all session participants. The clients then perform the appropriate action. We elaborate on playback synchronization among clients in Section 3.2.

3. IMPLEMENTATION

3.1 Client and Server Communication

Our client is written in JavaScript and uses SoundManager, a JavaScript audio player API. In contrast, our server code is written in Python 2.7. Both use the Google App Engine API.

When the client wants to contact the server, it issues an HTTP request, which is handled by the appropriate Python script handler. Handlers in Google App Engine can only serve requests within a time limit, which is why we chose to store session data within the Datastore.

In contrast, when the server wants to contact a client, it uses the Channel service. Channels are persistent connections that allow updates to propagate to clients without the use of polling. This is the mechanism with which we send updates to participants of a session. A channel is created as follows:

1. The client contacts the server
2. The server generates a channel ID and sends back a token
3. The client connects to the channel using the token

The client receives updates by listening on the channel by opening a socket and does appropriate actions based on the message received. Updates to the channel are made via HTTP requests from clients. When a client disconnects from the channel, a message containing the corresponding channel ID is sent to the server, and appropriate cleanup is performed.

3.2 Session Synchronization

In order to synchronize playback among different clients, we use control messages, which are propagated via channels as described above. Only hosts may issue commands such as play, pause, and next for the session. The “play” and “pause” commands are sent only when the host clicks on the pause and play buttons, and the “next” command is sent when the host switches songs or if the host clicks on the “next” button.

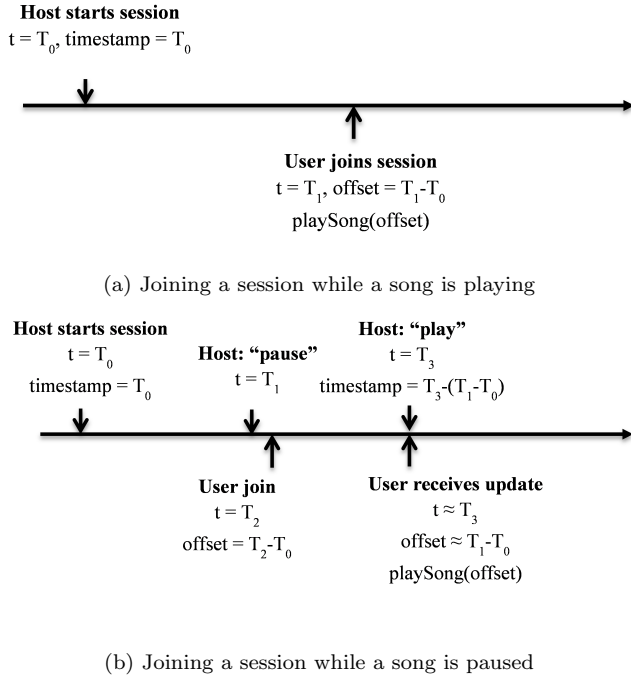


Figure 2: Synchronization messages when a user joins a session

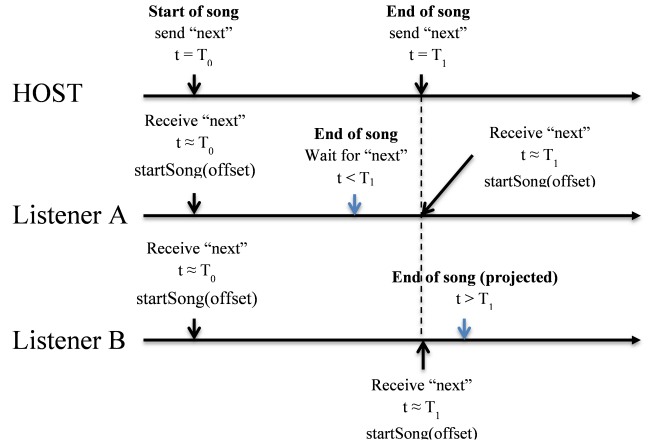


Figure 3: The synchronization timeline for a listener that’s ahead of or behind the host

We separate the problem of synchronization into two parts: joining an existing session and updating the current session. In both cases we utilize a timestamp, which is updated each time a play or next command is sent by the host client and is sent with the control message.

As seen in Figure 2 when a user joins a session, the server sends the user the elapsed time since the last stored timestamp. The user then start the song with the value returned by the server as the offset. If the song is playing when the user joins the session, it will start playing at the specified offset. If the song is paused when the user joins the session, the retrieved offset will not be the correct elapsed time; however, this is corrected when the host sends a “play” message to the server, which stores another timestamp and propagates the correct offset to all session listeners.

This timestamping synchronization is used mainly when a user joins a session. In the other case where participants update their current session upon receiving a control message, the offset can usually be inferred based on the local state. For instance, if the user receives a “next” message, then they start the next song with an offset of 0. If a user receives a “pause” message, they stop the current song and keep the elapsed time as the offset. If the user receives a “play” message they start the current song at the local offset, which was set when the “pause” message was received. The exception to this is if the current paused song is at the beginning, indicating that this user has joined the session while the song was paused or that this user is ahead of the host and is waiting for a control message from the host. In this case the user adjusts the current song offset with what they receive from the server. Conversely, if the listener lags behind the end of the song and start the next song. Figure 3 illustrates the timeline in the cases where a listener is behind the host and where a listener is ahead of the host.

3.3 Design Considerations

We made certain design decisions in order to achieve the goals discussed in Section 1. In this section we summarize and discuss these choices.

1. **Performance:** Songs are prefetched by the client for

seamless transitions. We also perform synchronization, as discussed in Section 3.2.

2. **Availability:** We rely on Google services such as Datastore and Blobstore, which are highly replicated and reliable. Requests are also served as long as Google App Engine is available. When a listener goes offline, this does not affect session availability of the other listeners.
3. **Scalability:** For scalability purposes, we allow concurrent requests. To minimize the number and size of messages, we allow incremental updates. We do not send all session data on an update

4. EVALUATION

We will evaluate our system on the metrics of performance, availability and scalability. To measure performance, we will design benchmarks to measure the data rate of a listener versus the data rate of the host. To evaluate availability, we will examine how our system behaves with various failure modes by artificially taking down each role (client, master server, session server). For scalability, we will measure system performance as we add clients to the system. We can achieve this by launching our application on an increasing number of client machines.

4.1 Performance

For the performance of our system, we are interested in differences in playback between clients in the same session. We want clients to be listening to not just the same song, but the same part of the song. To test this, we set up two different experiments.

1. Determine differences in playback between a host and a listener as more active sessions are added to the system.
2. Determine differences in playback between a host and a listener as more listeners are added to their session.

The host and listener used different users on the same machine so as to have identical clocks. The added sessions or listeners were achieved through a modified version of Google’s load test for App Engine applications. Differences in playback were observed through comparing the ending timestamps of songs between the host and listener. Each experiment followed a similar series of steps, listed below.

1. The host starts a session by choosing a song.
2. The listener joins the host’s newly created session.
3. 10 new sessions/listeners are added to the system.
4. The time the current song ended is recorded for both the host and listener.
5. The host adds a new song. Go to step 3.

The results for adding more active sessions are shown in Figure 4. As more sessions were added to the system, there wasn’t much of a difference in song playback between the host and listener. On average the difference actually decreased as more sessions were added, but this is likely due to the outlier observed for 30 sessions. The minor differences

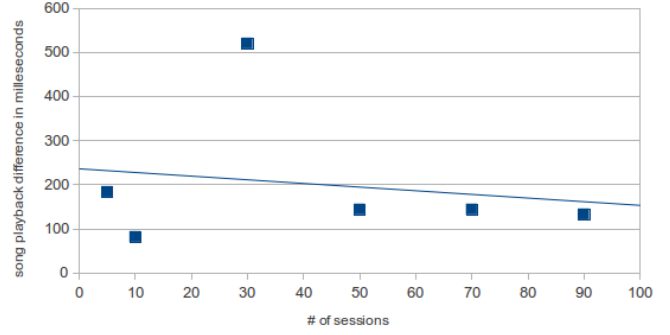


Figure 4: Playback difference when adding more sessions.

observed are more likely to have been caused by external network-related factors than our system itself.

The results for adding more listeners are shown in Figure 5. As more listeners were added to a session, there was a minor increase in song playback differences between the host and listener. However, even with 90 active listeners in a session, the difference was less than 1 second. This increase is likely due to the additional work the server has to perform to propagate updates to an increasing number of listeners in the session.

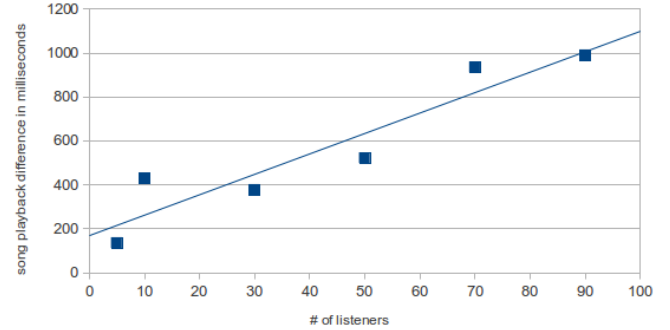


Figure 5: Playback difference when adding more listeners.

We also recorded the response time of our system as more listeners were added to a session. The results are shown in Figure 6. The response time increased as more listeners were added to a session, starting at around .5 seconds and reaching nearly 5 seconds. This is to be expected as the server propagates session updates to all listeners when a new listener joins a session. As more listeners join a session, the server has to send more updates. Even with nearly 100 listeners, the response time was still under 5 seconds. This is likely acceptable, but not ideal. To reduce this time, the server could propagate updates only on song changes which would reduce its overhead at the cost of hosts having an outdated view of session information.

5. FUTURE WORK

<ADD FUTURE WORK INFORMATION HERE>

6. CONCLUSION

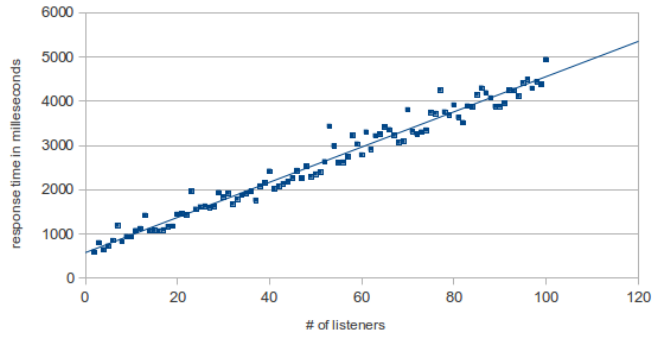


Figure 6: Response time when adding more listeners.

We will be developing a music sharing application called ClouDJ. With ClouDJ, users will be able to host music sessions that other users can listen to in real-time. It will consist of many client users as well as some centralized servers to handle session and user state. We will make use of Google App Engine for the implementation of the system.