CS 4455: Video Game Design & Implementation March 31, 2006: Audio

(Insert Disclaimer Here)

Overview

Today's Lecture What I'm talking about now Audio Theory Digitizing Sound Game Implementation High Level APIs

Why is Audio important?

What is audio?

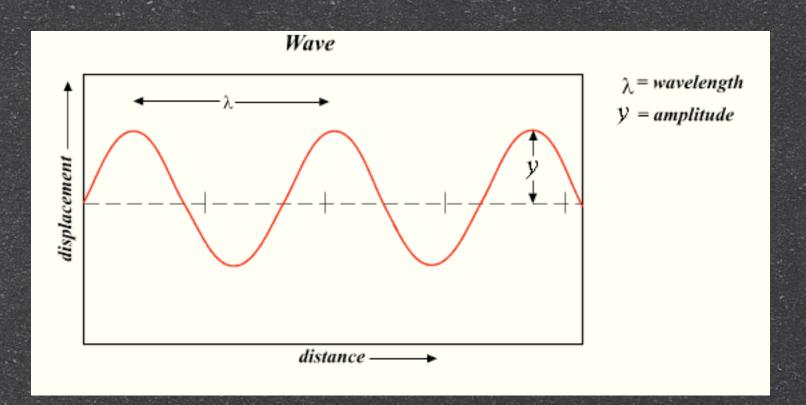
Inside your ear is an eardrum
 A thin piece of skin
 When it vibrates, your brain interprets this as sound
 Changes in air pressure often cause this vibration

How Audio Works

An object produces sound when it vibrates

This moves air particles

Those particles in turn move other particles

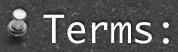


Terms to Note:

Wavelength - distance between repeating points

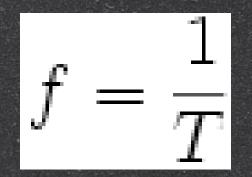
Amplitude - non-negative height of the wave

Audio Terminology

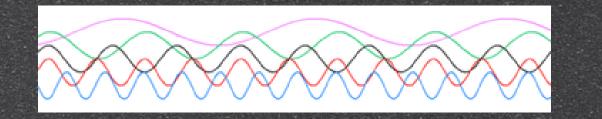


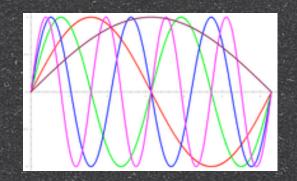
- Period How long it takes between cycles
- Frequency How many cycles occur





Audio Terminology





The faster they loop, the higher their frequency.

The SI unit for this is (Hz). 1 Hz = once a second, 1 KHz = one thousand times a second

Audio Terminology

Intensity = the "power" of the sound
A fairly large scale, so usually expressed logarithmically:

$$I_{\rm dB} = 10 \log_{10} \left(\frac{I}{I_0} \right) \quad {\rm or} \quad P_{\rm dB} = 10 \log_{10} \left(\frac{P}{P_0} \right) \ , \label{eq:IdB}$$

With Sound, IO ~= 10^-12 W/m^2

Common DB Levels

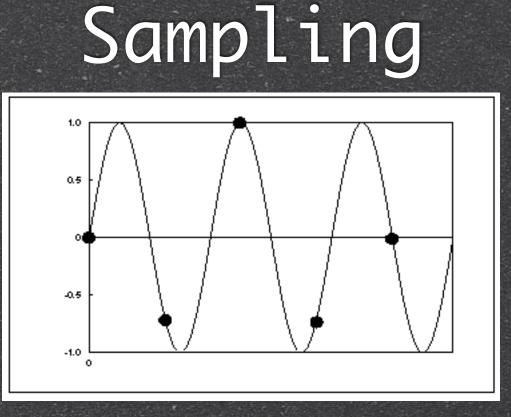
0dB	Threshold of hearing
10dB	Human breathing at 3 meters
30dB	Theatre, no talking
60dB	Inside of office or restaurant
70dB	Busy traffic at 5m
90dB	Loud factory, heavy truck at 1m
100dB	Jack Hammer at 2m; inside disco
120dB	Rock Concert
150dB	Jet engine at 3m
250dB	Inside tornado; nuclear bomb @ 5m

+10dB means 10 times as powerful +3dB roughly twice as powerful

Fun facts about audio strength

85dB	potentially harmful to hearing
120dB	unsafe
150dB	physical damage to body
163dB	windows break
19xdB	eardrums rupture
200dB	can cause death

Digitizing Sound



At a given interval, "sample" the amplitude of the wave

Sampling: Nyquist Limit

Nyquist Limit - a given sampling rate can only represent frequencies up to one-half that rate

Sampling

Typical factors on a computer
How many times per second?
How many levels can we differentiate between?
How many channels?

Sampling CD Quality audio 44kHz (44,100 samples/sec) 16-bit (65,536 possible levels) 2 channels (left and right) 176,400 bytes/sec This is approximately 6 seconds per megabyte!

Sampling

Low Quality Audio 8kHZ (8 samples per second) § 8-bit (1 byte) 1 channel 8,000 bytes/sec About two minutes per megabyte

Sampling Comparison



Sampling Comparison



What can we do? (Or, a brief history of computer audio)

FM synthesis

Used in early systems like the GameBoy

- Hardware continually produced one or more sine waves (sometimes other shapes, too)
- Software could modify frequency and amplitude

Can be done in very little space (the BIGGEST GameBoy games were about 1/2MB)

FM synthesis demo



MIDI synthesis

Common in PC sound cards and many consoles, such as the Super Nintendo (SNES), Genesis, your cellphone...

MIDI files contain instructions to turn on or off various instruments

Instruments are externally defined

Therefore, small file format

Sound can differ player-to-player

MIDI synthesis demo



Module Audio

Like MIDI, but you can (or are required to) supply your own instruments

.MOD/.S3M/.XM/.IT file formats

Used in the PlayStation, also common on the GameBoy Advance

Usually still small, if you can share instruments

Module Audio Demo



RedBook Audio

Music is streamed from CD
 Commonly used on the SegaCD
 May need to buffer or avoid when you need to read from the CD!
 Takes a lot of space

RedBook Audio Demo



(Realistic Depiction of Bonus Stage)

Compressed Audio

General Compression
Lossy - small changes okay
Loseless - must be 100% preserved

Compressed Audio

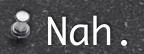
 Audio Compression
 Bit Reduction
 DPCM encodes the differences between subsequent samples (the D is for differential or delta)
 ADPCM is a more advanced version

Compressed Audio

Audio Compression
 Psycho-acoustic
 Designed with human hearing in mind

MP3, AAC, OGG, WMA, .etc

Compressed Audio Demo



Gaming Implementation

Gaming Implementation

- Sound in most games is divided into two parts:
 - Background music
 - Sound Effects

Background Music
Unlike movies and TV, not timed
Ambient and looping
Can be streamed



Sound Effects

Characteristics Typically very short Often tied to an event Examples: Gun fire, character is hit, explosion, speech, .etc Generally stored in memory

Mixing

In many game consoles, separate HW for these functions is not unusual
 Hardware MIDI standard in some
 If not, convert to sampled in software

Mixing

Software mixing is easy
 If sample rates are the same
 Just add!
 Beware of exceeding the max

Mixing Example

def mixSound(dest, source):
 for i in range(1, min(getLength(dest), getLength(source))):
 sourceValue = getSampleValueAt(source, i)
 destValue = getSampleValueAt(dest, i)

setSampleValueAt(dest, i, sourceValue + destValue)

Buffering

When reading or converting sound, you need to stay ahead of the audio out device but can't convert the whole song

- Two techniques for buffers:
 - Circular buffers read and write in same buffer
 - Buffer chaining write to buffer, read from the other, swap

High Level APIs

- Audio is a lot simpler than graphics (in a game).
- A lot of APIs can be condensed to:
 - Play(sample, loops)
 - Stop(sample)
 - SetPan
 - SetVolume
 - SetSpeed

Example Sound APIs

Cross platform 2D APIs 🗳 Java Sound SDL QuickTime Sound APIs with 3D support OpenAL DirectSound (NOT crossplatform)

Example (JavaSound)

Sequence sequence =
 MidiSystem.getSequence(new java.net.URL(url));

// Create a sequencer for the sequence Sequencer sequencer = MidiSystem.getSequencer(); sequencer.open(); sequencer.setSequence(sequence); sequencer.start();

See packages under javax.sound such as javax.sound.midi and javax.sound.sampled

Positional Audio

Basic Theory

- Sound distance and volume are inversely related
- Sound differences in the ears help determine position

3D Audio

Problem with traditional two speaker or headphone setup

- Forward vs. Behind
- Speaker setups available that have 5, 6, or 7 speakers

Speaker setup tends to differ, so this is difficult!

3D Audio Example

// Load wav data into a buffer.

alGenBuffers(1, &Buffer);

if (alGetError() != AL_NO_ERROR)
 return AL FALSE;

alutLoadWAVFile("wavdata/Footsteps.wav", &format, &data, &size, &freq, &loop); alBufferData(Buffer, format, data, size, freq); alutUnloadWAV(format, data, size, freq);

// Bind buffer with a source.

```
alGenSources(1, &Source);
```

if (alGetError() != AL_NO_ERROR)
 return AL FALSE;

alSourcei (Source, AL_BUFFER, Buffer); alSourcef (Source, AL_PITCH, 1.0f); alSourcef (Source, AL_GAIN, 1.0f); alSourcefv(Source, AL_POSITION, SourcePos); alSourcefv(Source, AL_VELOCITY, SourceVel); alSourcei (Source, AL_LOOPING, AL_TRUE);

// Do an error check and return.

```
if (alGetError() != AL_NO_ERROR)
    return AL FALSE;
```

```
// Initialize OpenAL and clear the error bit.
alutInit(NULL,0);
alGetError();
```

```
// Load the wav data.
if (LoadALData() == AL_FALSE)
    return 0;
```

SetListenerValues();

// Begin the source playing.
alSourcePlay(Source);

```
// Loop
ALint time = 0;
ALint elapse = 0;
```

```
while (!kbhit())
```

```
elapse += clock() - time;
time += elapse;
```

```
if (elapse > 50)
```

```
elapse = 0;
```

```
SourcePos[0] += SourceVel[0];
SourcePos[1] += SourceVel[1];
SourcePos[2] += SourceVel[2];
```

```
alSourcefv(Source, AL_POSITION,
```

```
SourcePos);
```

Resources

OpenAL tutorials:

http://www.devmaster.net/articles/
openal-tutorials/lesson1.php

Sound Editing Software:

http://audacity.sourceforge.net/

Sound Effects

/net/dvfx/hollywood_edge

http://www.sounddogs.com/