CSE481i: Digital Sound Capstone

An Overview



(Material freely adapted from sources far too numerous to mention...)

Today



- What this course is about
 - Place & time
 - Website
 - Textbook
 - Software
 - Lab
 - Topics
- An overview of Digital Sound Engineering

Where and when



- We will meet Tuesdays and Thursdays, 10:30-11:50 p.m. in cse 203.
- Lab is Tuesday 2:30-5:20 in CSE 003E
- This week:
 - Check your card access and login in the lab.

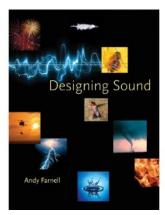
Recommended Textbook



RECOMMENDED textbook:

Designing Sound Andy Farnell The MIT Press

NOT Required, but recommended Some pdf excerpts are available



Other Useful Books



The SuperCollider BookScott Wilson, David Cottle, Nick Collins and James McCartney

The Audio Programming Book

Edited by Richard Boulanger and Victor Lazzarini

The Computer Music Tutorial

Curtis Roads

Real Sound Synthesis for Interactive Applications

Perry R. Cook

Unity Game Development Essentials

Will Goldstone

More on the website

Software



Major packages we'll use:

Pure Data

graph-based sound language public domain

SuperCollider

client-server based sound synthesis language

The Synthesis ToolKit in C++ (STK)

physical modeling toolkit

Audacity

Sound editor

Kinect API

Human Sensing

The Lab- CSE 003E

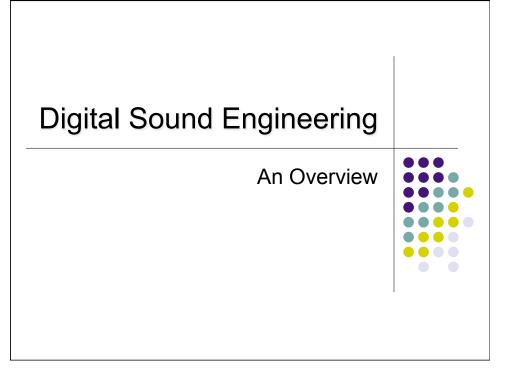


- This room is available for your use 24 hours a day
- My office hours are the lab times; you can visit me anytime in CSE 464
- You can do your work anytime, but the lab is where you can get help

Tentative topics



- Introduction and overview
- Fundamentals of Acoustics
- Fundamentals of Digital Audio
- Oscillators
- Modulation
- Sampling
- Non-linear Synthesis
- Granular Synthesis
- Sampled sound for animation
- Human sensing—Kinect API, Leap Motion API
- Presentations
- Capstone project planning









Or this:





Or this:





Or this:





iPad Apps for Music Making:



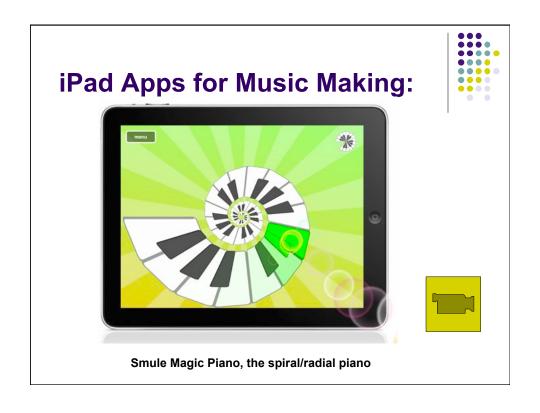


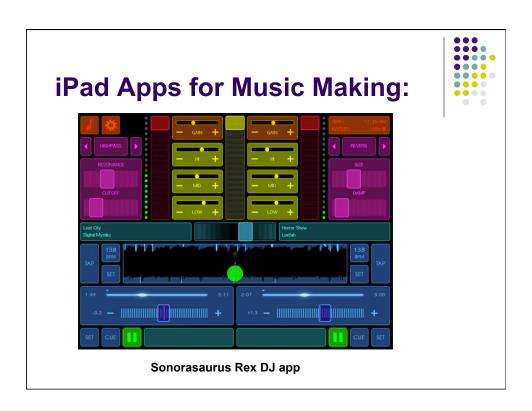
Sonoma's StudioTrack













So, how does the iPad make music?

Or, How do we produce sound with a computer?



Computer Audio



- An interdisciplinary field including
 - Music
 - Computer Science
 - Electrical Engineering (signal processing)
 - Physics (musical acoustics)
 - Psychology (psychoacoustics, music perception)
 - Physiology (hearing, speech and singing)

Computer Audio Taxonomy



- Signal Processing
 - Sound Analysis and Resynthesis
 - Physical Modeling of Musical Instruments and Speech
 - Musical Effects
 - 3D Spatialization
 - Audio Coding and Compression
 - Audio Signal Separation
 - Music Signal Pitch Detection
- AI
 - Machine Recognition of Audio and Music
 - Musical Instrument Recognition
 - Music Perception and Cognition
 - Psychoacoustics
 - Al and Music

Computer Audio Taxonomy



- Software
 - Music Visualization
 - Music Composition Systems and Tools
 - Music Programming Languages
 - Algorithmic Composition
 - Music Notation and Printing
 - Music on the Internet
 - Music in Computer Games
 - Sound Effects in Computer Games
 - Computer Music and Digital Art
- Database
 - Music Information Retrieval
 - Musical Informatics
 - Music Databases

Computer Audio Taxonomy



- Computer Engineering
 - Audio Hardware
 - Music Performance Interfaces (new musical instruments)
 - Interactive Performance Systems
 - Real Time Performance Systems
 - Music Workstations
 - Soundcards
 - Music Synthesizers
 - Music and Audio on Mobile Phones
 - Wireless Audio Systems
 - Music Networks
 - MIDI

Computer Audio Taxonomy

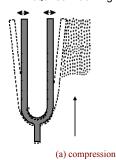


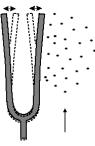
- Theory/Science
 - Music Data Structures and Representation
 - Musical Tuning and Intonation
 - Music Grammars
 - Musical Acoustics
 - Acoustics of Musical Instruments and the Voice

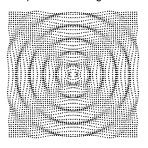
What is Sound?



- Variation in air pressure caused by compression and decompression of molecules
- Caused by friction producing force (stick striking symbol, finger plucking guitar string)
- · 'Waves' produced by cohesion of molecules, which fall on eardrum or microphone
- Directly and through reflection off surfaces in room
- Ear can detect frequencies in the range 20Hz to 20khZ
- Ear has very high dynamic response compared with eye (ie ability to detect changes in pressure)
- · Requires much higher sampling rates to digitize audio compared with images







(b) rarefaction

(c) wave propagation of a tuning fork as seen from above

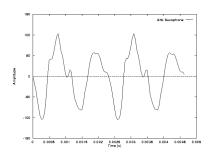
Properties of sound...

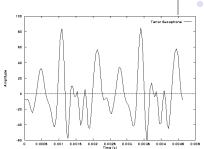


- Waveform distinctive pattern of variations in air pressure
 - Musical instruments produce orderly repeating waveforms
 - Noise produces random chaotic waveforms
- Fourier demonstrated how any wave form can be decomposed into a series of component sine waves of different frequencies
- Different frequency components, or pure tones, which are added together to produce a complex waveform are called the frequency spectrum of that waveform

Same note.. different waveforms







- Both figures show an 'A' note, left played on an alto sax and the right on a tenor sax.
- Both have additional frequencies as well as the main 440Hz

Physical and subjective attributes..



- Important to distinguish between the properties of a stimulus and those of a subjective response to that stimulus
- A linear increase in the stimulus value does not necessarily produce a similar increase in the subjective response

Stimulus value	Subjective response
(luminance)	(brightness)
Amplitude of wave	Loudness of sound
Frequency of wave	Pitch of sound
Several attributes (hard to define)	Timbre of sound

Amplitude and Frequency



- Amplitude measured in decibels
 - The louder a sound is, the more it will mask or dominate other other sounds adjacent to it in time
- Frequency measured in cycles per second (Hertz Hz)
 - More digital information required to encode higher frequency sounds, lower pitched sounds degraded less by low sample rates

Timbre

- loosely defined by 'tone', 'color', 'texture' of sound that enables brain to differentiate one tone from another
- Affected by acoustic properties of instruments and room

Digitizing sound



- Analog signal is sampled and converted to a series of digital values (A to D converter)
- Digital values later converted back to analog for playback through speakers (D to A conversion)
- Parameters are frequency at which samples are taken and the resolution of each sample (i.e number of bits used to encode analog signal value
- Nyquist's theorem prescribes minimum sample rate in order to be able to re-construct analog signal
- If maximum frequency in the waveform is n Hz, then minimum sample rate should be 2n Hz

Sampling and Quantizing





- Sampling process of acquiring an analog signal
- Quantizing conversion of held signal into sequence of digital values

Sample rates



- If upper range of ear is 20Khz, then there is no need to faithfully reproduce frequency components in signals higher than this.
- CD quality: at least 2 x 20KHz = 44.1KHz
- The human voice has few frequency components lower than 100Hz, or higher than 3000Hz - a bandwidth of 2900Hz
- **Speech**: at least 2 x 2.9KHz = 8KHz

Sample data rates



- For CD quality,
 - Rate = 44.1Khz (44100 samples per second)
 - Resolution = 16 bits
 - Stereo = 2 channels
- Data rate = 44100 * 16 * 2 bits/second = 1411200 bits/sec
- (10Mb storage for 1 minute of recorded sound)

Examples of data rates and quality Stereo/Mono Sample Rate Resolution Bytes (1 min) 44.1 KHz 、 16 bit Stereo 10.1 Mb 44.1 KHz & bit Mono 2.6 Mb 16 bit 22.05 KHz Stereo 5.25 Mb 22.05 KHz 8 bit Mono 1.3 Mb 11 KHz & bit Mono 650 Kb 5.5 KHz 8 bit Mono 325 Kb CD quality audio As good as a TVs audio As good as a bad phone line

Digitized vs. Synthesized



Multimedia sound comes from two sources:

- Digitized from an external (sampled) real life sound
- Synthesized created from waveforms in a sound card for example

Traditional analog sound synthesis is achieved by

- Creating a waveform using an oscillator, which sets the basic frequency
- Adding an "envelope", by specifying parameters such as attack, decay, sustain, release
- Then sending through filter(s) to modify timbre

MIDI - Musical Instruments



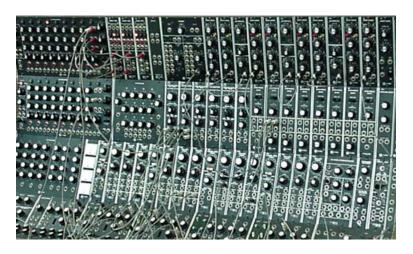
- Digital Interface supported by many instruments/ computers/ manufacturers (1980)
- Defines set of messages indicating note/ instrument/ pitch/ attack etc
- Sound card/ Synthesizer takes this symbolic message and 'creates' matching sound
- Sampled sounds can be stored by users on better equipment
- Compare waveforms to bitmapped images, midi to vector graphics

Digital Sound Synthesis Methods



Depending on your age, you might think the first synthesizer looked something like this:





However, it looked more like this:





Synthesis Definition



The Oxford Classical Dictionary defines synthesis as:

- Combination, composition, putting together
- Building up of separate elements into connected whole

Synthesis Definition



- Generally, most people associate synthesis purely with <u>subtractive</u> synthesis
- Very limiting way to look at sound synthesis by electronic means

The Bigger Picture

- Theoretically sound divisions, but practically limiting
- Techniques of different types applicable to others

Subtractive Sampling Analog Granular Waveshaping FM Physical Modeling Additive

In the beginning...



- Additive synthesis
- Principle first utilized in cathedral organs:



Additive Synthesis

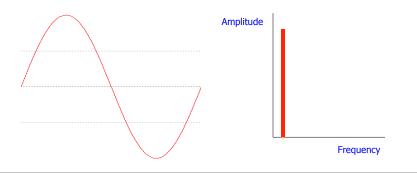


- Mathematical basis:
 - 1822: Jean Baptiste Joseph, Baron de Fourier published theory:
 - Any arbitrarily complicated periodic waveform can be deconstructed into combinations of sine waves of different amplitudes, frequencies and phases
 - -This is accomplished by the *Fast Fourier Transform*: FFT

Additive Synthesis



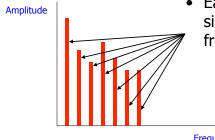
- Sine wave = simplest possible waveform
- Contains only the fundamental



Additive Synthesis



 A more complex waveform will be composed of any number of sines of varying frequencies and amplitudes:



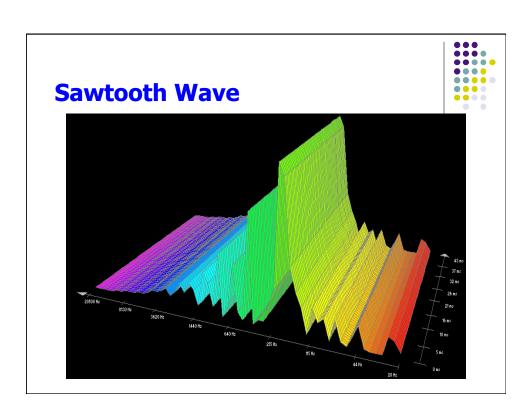
Each line represents a sine at a specific frequency and amplitude

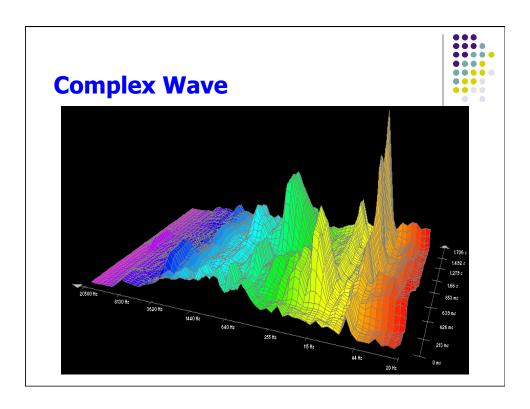
Frequency

Additive Synthesis



- But this simple approach hides many difficulties
- Theory shown so far deals with a single moment in a sound's duration
- Most sounds are complex and evolving

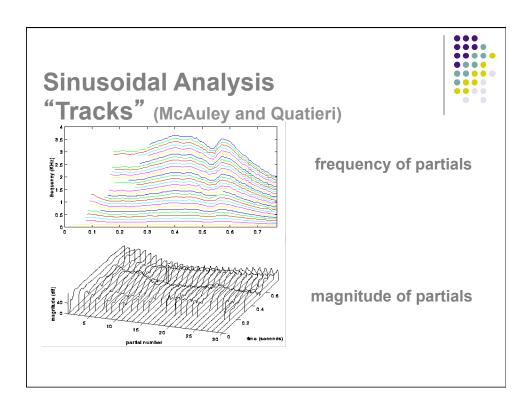


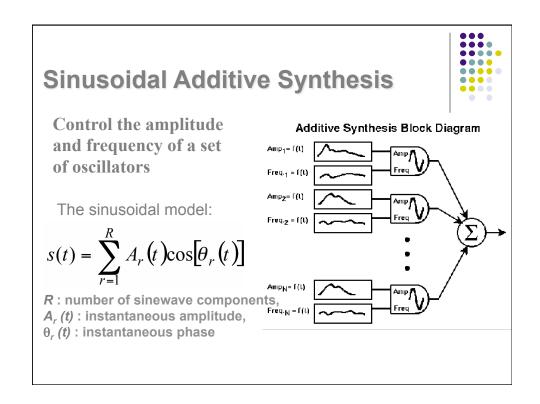


Additive Synthesis



- Thus, will have multiple slices depending on:
 - Length of waveform
 - Rate of change of waveform
- Control data therefore massive
- Very hard to create sounds using additive synthesis
- Holy Grail: Analysis-Based Resynthesis





The Bigger Picture

- Theoretically sound divisions, but practically limiting
- Techniques of different types applicable to others

Subtractive

...

Sampling

Analog

Granular

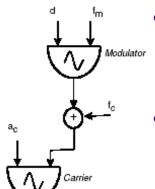
Waveshaping

FM

Physical Modeling

Additive

FM Synthesis



- Simple FM: carrier oscillator has its frequency modulated by the output of a modulating oscillator.
- Sidebands produced around carrier at multiples of modulating frequency.
 - Number generated depends on the amplitude of the modulator.



FM Synthesis



• Nothing more than an extreme form of vibrato:



- When the modulation is fast enough, we no longer hear the rise and fall of the vibrato
- Instead, we perceive the changes in pitch as changes in the timbre of the sound

Modulator: Carrier Ratio

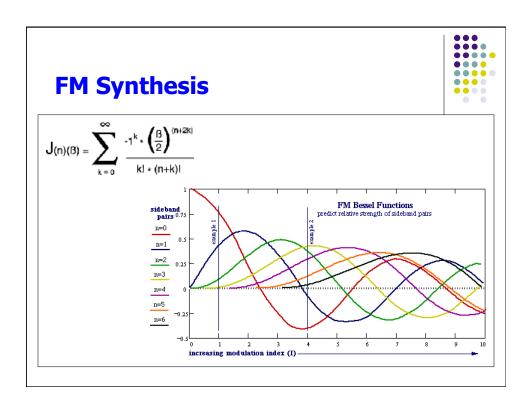


- Sidebands at C + and (n * Modulator)
- Ratio of M:C determines whether spectrum is harmonic or not.
 - Simple integer ratio = harmonic
 - Non-integer ratio = inharmonic



Modulation Index and Bandwidth

- The *bandwidth* of the FM spectrum is the number of sidebands present.
- The bandwidth is determined by the Modulation Index
 - I = depth of modulation / modulator
 - D depth of modulation, which depends on the amount of amplitude applied to modulating oscillator. ($D = A \times M$)
- If the index is above zero, then sidebands occur.



FM Synthesis



- Unfortunately, the relationship between these is not predictable without experience:
 - as the Index changes, the amplitude of each sideband pair evolves in a different pattern
 - some sidebands gain amplitude, others lose amplitude
 - there may also be cancellation effects caused by phase-inverted sidebands.
- This remains the most significant barrier to learning FM synthesis
- Nevertheless a powerful technique for creating complex sounds

Granular Synthesis

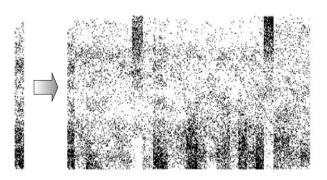


- Attempt to deal with the shortcomings of additive synthesis to deal with changes in the sound over time
- 1947: Dennis Gabor, physicist formulated theory:
 - sound is perceived as a series of short, discrete bursts of energy, each slightly changed in character from the last
- Rooted in quantum physics coexistence of the wave and photon in light
- Sonic equivalent of the photon is the grain

Granular Synthesis



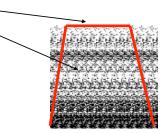
- *Definition*: generation of thousands of short sonic grains which are combined linearly to form large scale audio events
- Grain = tiny piece of sonic data, duration: 10 to 50 ms.



Granular Synthesis



- Two components:
 - Envelope -
 - Contents <



- NB: Grain Density number of grains per second
 - Low density leads to rhythmic effects

Granular Synthesis



- Subject to same fundamental problem as additive synthesis, though:
 - Tension between precision and control
 - Massive number of grain events
- Basic unit -> grain cloud rather than grain itself
 - ~ Set of rules for generating and controlling grains
- It has some of the drawbacks of FM synthesis as well:
 - Unpredictable results
- But capable of creating sound textures that no other form of synthesis can

Subtractive Synthesis

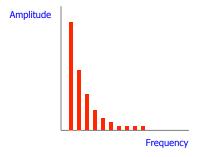


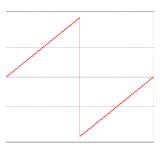
- Well understood and widely employed
- Begin with a harmonically rich sound source and remove frequencies by means of filtering
- While any sound source can be employed, traditionally associated with certain waveshapes

Subtractive Synthesis



• Sawtooth: contains all harmonics, with amplitude 1/n:

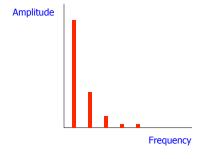


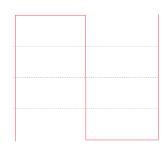


Subtractive Synthesis



• Square: only odd harmonics present, also with amplitude 1/n

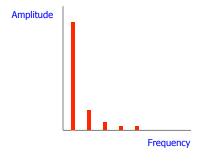


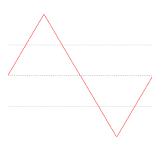


Subtractive Synthesis



 Triangle: only odd harmonics present, but with amplitude 1/n²

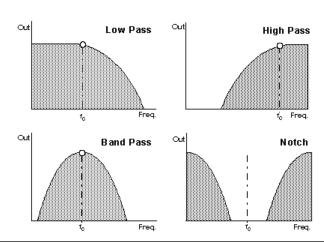




Subtractive Synthesis



• Basic Filters



Waveshaping Synthesis



- Sound of a waveform determined primarily by its harmonic content
- Can create new harmonics by passing waveform through non-linear element: waveshaper
- Often a Chebyshev polynomial

Physical Modeling



- Modeling sound generation
 - more expressive and realistic sounds
 - ideal for software implementation
 - no need for dedicated hardware

• Brute force approach

- solve equations of motion with respect to boundary conditions
- Better Approach
 - Partway solve equations for changing parameters
 - lookup tables
 - lumped processes
 - novel algorithms



Yamaha VL1

Digital Wave-guide Modeling

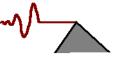


Da-lembert's Solution .. Waves travel in equal an opposite directions

$$f(ct+x)+g(ct-x)$$

$$\longrightarrow$$

Reflected and attenuated at boundary



Boundary behaviour frequency dependant



Digital Wave Guide Modeling

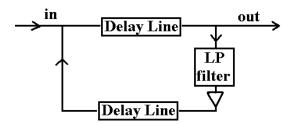


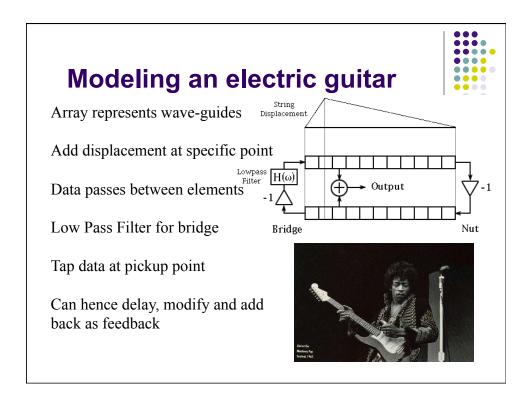
Energy in (pluck, hammer, bow, etc)

Delay line simulates time wave travels

Feed back loop simulates reflection

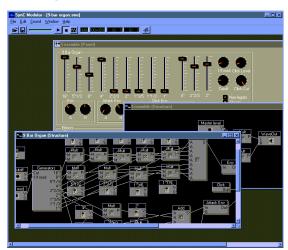
Filter simulates frequency dependant attenuation





Modular and Virtual Modular Synthesis





SynC modular

A modular paradigm allows for additive, subtractive, fm and sampling synthesis techniques to be used together