CS 330: Homework 3

Nullaby

Overview

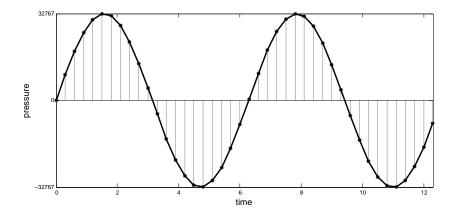
Your task in this homework is write a tool for converting an XML representation of a melody written in the standard Western musical scale to a WAV file. You will digitally synthesize each note. Because there is no way to predict how much memory will be needed ahead of time, you will dynamically allocate large chunks of memory. When all the samples of all the notes have been stitched together, the melodious stream can be written to disk.

As you complete this homework, you will gain more experience with pointers, dynamic memory allocation, binary files, protocols, higher-order functions, push parsing, callback-driven systems, and the C++ programming language.

Digital Sound

Sound is a wave of pressure that propagates through the physical world. How fast a sound wave moves determines its pitch. This rate of movement, or *frequency*, is measured in cycles or oscillations per second.

We record sound using some physical device that is sensitive to a wave's changing force. Early analog recorders used a vibrating stylus to etch a continuous alternate representation of the sound into foil, wax, or paper. Today, recording is done with microphones that produce an electrical signal as they vibrate. Modern recording equipment is digital—capturing only a countable sampling of the wave's force. Pressure readings are relative to the microphone's resting position. When the pressure is greater than equilibrium, we see positive values. When the pressure is lower, we see negative values. If we plot these readings over time t, we see a displacement graph like the following:



One of the beautiful things about this digital model is that we can synthesize sound artificially—we don't need to record something first. In fact, to create a sound clip for a given frequency and volume, we simply walk along the wave represented by amplitude $\sin(2\pi t \cdot \text{frequency})$. We capture the magnitude of the wave at all sample times t and write them out to disk using some digital sound protocol like the WAV format. Our sound card and speakers will convert these samples back into a continuous signal that is blasted toward our eardrums.

XML Music

The musical notation for this homework is a simple XML format. The following example demonstrates a possible (i.e., random and terrible) melody containing notes of several durations and some repetition of note sequences:

Believe it or not, you don't need to have any background in music to be able to complete this assignment.

The following rules describe the XML file structure in more detail. You may assume the XML file is well-formed, but you cannot assume the attributes to be in any particular order.

- 1. The root entity is song, which you may assume has a default octave attribute, a default note/rest duration attribute, and a beats-per-minute attribute. Durations are of the form 1 (a whole note), 2 (half), 4 (quarter), 8 (eighth), 16 (sixteenth), or 32 (thirty-second).
- 2. Nested in song is a sequence of note, rest, and repeat entities.
- 3. All notes have a name attribute, one of A, B, C, D, E, F, G, optionally followed by a + (sharp, raise the note a half step) or (flat, lower the note a half step). Notes may or may not have octave and duration attributes. If they do not, the containing song's attributes are used.
- 4. Rests may or may not have a duration attribute. If they do not, the containing song's duration attribute is used. Rests are effectively soundwaves with frequency 0.

5. Repeated sequences are grouped in a repeat entity. You may assume such entities have a times attribute indicating how many times the sequence is to be inserted in the song. If times is 2 and the contained sequence is ABC, the repeated sequence effectively expands to ABCABC. You may assume the times attribute is greater than 0. Repeated sequences may nest.

Requirements

Specifications do not tell you how to solve a problem—just what pieces may be used. The requirements below and the descriptions above will need to be thought about and pieced together using your own good mind. You will likely need to read this document many times, ask questions, and afford yourself plenty of time to finish this homework.

Your solution is to meet the following specification:

- Place all files in directory <YOUR-REPOSITORY>/nullaby.
- All code must run on dplsubmit.
- Write code in C++. You may use functions from either the standard C or C++ libraries.
- Keep functions and methods short, document any code chunk whose purpose isn't immediately obvious, and use meaningful names.
- Declare each class in its own header file and implement it in its own implementation file. Name the files using the class name, e.g., Note.h and Note.cpp.
- Write a class Pitch with the following specification:
 - Has a constructor that accepts a duration parameter as an int.
 - Has a virtual destructor. (See http://www.parashift.com/c++-faq/virtual-dtors.html.) Likely this method need do nothing but exist.
 - Has a pure virtual const method named GetFrequency that returns the pitch's frequency as a float.
 - Has a non-virtual const method GenerateSamples that accepts as parameters a beats-per-minute as an int and a number of samples as an int reference. It returns a pointer to a short buffer containing samples for the pitch. The number of samples is stored in the reference parameter. A clue as to how you might generate the samples is shared below. What the reference parameter holds when this method is called is undefined. It is an out parameter.
 - Has a pure virtual const method named Clone that returns a pointer to a dynamically-allocated copy of this pitch.
- Write a class Note with the following specification:

- Has a constructor that accepts as parameters the note's name as a const string reference (matching /^[ABCDEFG] [+-]?\$/), its octave as an int, and its duration as an int.
- Has a const method GetHalfStepID that returns the note's halfstep number as an int. Consider the halfstep IDs of all possible notes in the first octave:

C0	C+0/D-0	D0	D+0/E-0	E0	E+0/F0	F+0/G-0	G0	G+0/A-0	A0	A+0/B-0	B0/C-1
0	1	2	3	4	5	6	7	8	9	10	11

The IDs of the second octave are offsets of the first:

ſ	C1	C + 1/D 1	D1	D 1/E 1	E-1	Tr 1 /Tr1	F 1/C 1	C1	G+1/A-1	Λ1	A 1/D 1	D1/C1
- 1	10	1.0	1 4	1 -	10	1 7	10	10	20	0.1	00	0.0
	12	13	14	15	10	1 17	18	19	20	21	22	23

And so it goes. Note the octave starts at C, not A. (Hint: you can write this function without any loops or monstrous arrays. Use the letter of the note as an index into a baseline 7-element array.)

- Overrides GetFrequency to return the note's frequency. Physicists tell us to calculate the frequency of note X using these equations:

frequency of A4 = 440 $magic\ number = 2^{\frac{1}{12}}$ distance of X from A4 = halfstep ID of X – halfstep ID of A4 $frequency\ of\ X = frequency\ of\ A4 \cdot magic\ number^{distance\ of\ X\ from\ A4}$

- Overrides Clone to return a pointer to a dynamically-allocated copy of this note.
 Exploit the fact that we can have covariant return types, i.e., return the pointer as a Note *—not a Pitch *.
- Write a class Rest with the following specification:
 - Has a constructor that accepts as a parameter the rest's duration as an int.
 - Overrides GetFrequency to return the rest's frequency—which is 0.
 - Overrides Clone to return a pointer to a dynamically-allocated copy of this rest.
 Exploit the fact that we can have covariant return types, i.e., return the pointer as a Rest *—not a Pitch *.
- Write a class Nullaby that encapsulates a melody, which is a sequence of pitches. It has the following specification.
 - Has a static const member named SAMPLES_PER_SECOND with value 22050.
 - Has a default constructor.
 - Has a deconstructor that releases all resources owned by the melody.

- Has a const method GenerateSamples that accepts as parameters a beats-perminute as an int and a number of samples as an int reference. It returns a pointer to a short buffer containing samples for the melody. The number of samples is stored in the reference parameter. The samples are generated by concatenating all the pitches' samples together into a contiguous, all-encompassing short buffer.
- Has a method WriteWAV that accepts as parameters beats-per-minute as an int and path to a WAV file as a const string reference. It writes the melody's samples out to the specified file, conforming to the WAV specification described at https://ccrma.stanford.edu/courses/422/projects/WaveFormat. Use just one channel, select PCM, sample at 22050 samples per second, and write the samples out as shorts.
- Has a const subscript method that accepts an int index parameter and returns the pitch at the specified index as a const Pitch reference.
- Has an Add method that accepts a const Pitch reference as a parameter. It
 adds a clone of the pitch onto the end of this melody. Bear in mind that since
 Nullaby triggered the allocation of the clone, it is responsible for releasing the
 clone's resources when finished with it.
- Has an Add method that accepts a const Nullaby reference as a parameter. It adds clones of all the pitches in the specified melody to the end of this melody.
- Has a const method GetPitchCount that returns as an int the number of pitches that have been added to this melody.
- Has a main function and all non-class code in main.cpp.
- Expects command-line arguments for the input XML file and a path to the output WAV file.
- Has a makefile whose default rule builds an executable named nullaby. Add a clean rule that deletes your executable. Compile with -g to retain more debugging information. Compile with -std=c++0x to use newer C++ features.
- Parses the XML file with the expat library. Use no global variables. Use XML_SetUserData to pass data around to your callbacks.
- Frees all dynamically-allocated memory. Valgrind must report no memory leaks.
- Is accompanied by an XML song that you create. Melodies need not be original. Share it on Piazza in a note tagged nullaby.

Generating Samples

Consider the following pseudocode for generating samples for a note of a given frequency and duration (which is 1, 2, 4, 8, 16, or 32):

```
beats_per_second = beats_per_minute / 60
seconds_per_beat = 1 / beats_per_second
seconds_per_whole_note = seconds_per_beat * 4 // assumes 4/4 time

nseconds = seconds_per_whole_note / duration
nsamples = nseconds * samples_per_second

make samples array

// i is an index, which we must convert to time
cycles_per_sample = frequency / SAMPLES_PER_SECOND
for each sample i
    sample i = sin(2 * pi * i * cycles_per_sample) * 32767
```

This only generates samples for an individual note. When you join them with samples from another note, you will hear some popping in the audio. I recommend you gradually increase the magnitude of the first seconds_per_whole_note / 64 seconds of each pitch from 0 it its full value. Likewise, diminish the last seconds_per_whole_note / 64 seconds from its full value to 0.

Submission

This homework is graded by hand and with help from the grading script, which can run from your homework directly with ../specs/grade. Your assignment is expected to fully meet the requirements above, those checked by the grading script, and the following:

- 1. Variable names should be meaningful and accurate.
- 2. Non-obvious parts of your code should be commented.
- 3. Code should be cleanly formatted and indented.

Your work will also be inspected for plagiarism. Please do your own work. Talk about code with your classmates. Ask questions of your instructor or TA. Do not look at others' code. Do not ask questions specific to your homework anywhere online but Piazza. (If you find violators of this rule, please let me know.) Write your own code.

The grading script sends your instructor an email when it successfully completes. All that remains is for you to push your code to Bitbucket.