

Parameter automation for granular synthesis

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Contents

1	Introduction	6
1.1	Aims and Objectives	6
1.1.1	Deliverables	7
2	Literature Review	8
2.1	Problem background	8
2.1.1	Audio descriptors	9
2.1.2	Granulation	9
2.1.3	Machine Learning	10
3	Methods	11
4	Results and Analysis	13
5	Discussion	14
5.1	Summary of Findings	14
5.2	Evaluation	14
5.3	Future Work	14
6	Conclusion	16

List of Figures

3.1 Basic flowchart 11

List of Tables

Chapter 1

Introduction

1.1 Aims and Objectives

The initial aim of the project was to implement a machine learning solution to the task of granular synthesizer programming based on sound matching. Consequently building a tool that would assist musicians in creating interesting sounds, provided an audio input to the system.

Over the course of the past couple months the project extended into more research-oriented directions, with a focus of finding the best possible audio descriptors for specific characteristics of sounds. In order to determine the most significant in judging similarity on the cognitive level.

The methods here concerned with describing audio differ from other approaches in literature (e.g Matthew) primarily in that sound is treated in a modular manner, where a description is made based on one characteristic for X number instances, and separate processes are run for each. Instead of trying to describe audio as a whole, the aim is to describe it as a combination of things (density, pitch, rhythm, amplitude)

Concretely my objectives are:

1. To build a tool that is helpful to artists in creating new sounds
2. To challenge the interaction between an artist and a preset as a starting point to synthesis
3. To achieve a response that is not only stimulating to the user but also differs from simply randomizing the parameter values

In measuring the project's success, the subjective sonic coherence and similarity of algorithm's

outputs may be considered the best indicator (the human discriminator?), along with more quantitative analysis, such as comparing the audio descriptors on input and output sounds for example.

In (future chapters) I provide some critique on established techniques of assessment of these types of problems and ultimately conclude that some stuff in the best way of doing it.

1.1.1 Deliverables

to these ends hehhe this is what i made:

Some stuff about things. Some more things. Inline citation:

Chapter 2

Literature Review

2.1 Problem background

Programming synthesizers is a fun, but challenging task. The range of sounds possible to achieve on such a device, and consequently the amount of adjustable parameters can be overwhelming. From choosing an audio file to sample, to an amplitude envelope for each grain, the ability to make decisions about programming a granulator has to come from either a place of certainty about what each parameter is responsible for, or a place of experimental thought, and a somewhat random parameter value assignments.

Users fairly new in the realm of synthesizer programming may encounter issues creating sounds they desire. Even successful musicians might be doing things following a not clearly defined “intuition”. There are of course people who are experts in this field, but artists often look for inspiration when it comes to timbre of their sounds. It would seem that, the use of presets as a starting point is not unusual.

Research has been done previously as an investigation into automation of parameters in synthesizers based on sound matching. Taking a snippet of sound, the algorithm would try to find parameter settings to match a produced sound as closely as possible to the source

This research mostly focuses on FM synthesis , although experiments on different synthesis techniques has been done , including any VST plug-in

Also, it seems that most of this research focuses on reproducing the original input . Perhaps more interesting and novel sounds could arise as an effect of bad performance, but it does not seem to be the desired outcome in most cases.

However, using this approach on corpus-based synthesis remains an untapped area of research, worth investigating

2.1.1 Audio descriptors

MFCCs are a very popular descriptor used widely in the domain of speech synthesis. Some literature also uses it as the means of comparing two sounds together (cite matthew, or some other study). It is probably the most sophisticated algorithm for describing timbre available currently.

There is of course a huge variety of audio descriptors available, and some serve different purpose than others. One can get very specific and use descriptors for onsets, pitch, etc.

These really gain power when used in parallel, or together, rather than on their own. That is the main reason behind the approach of describing audio in this project.

Especially when trying to describe audio to a machine - an agent of you will, that will be responsible for the programming of the granulator. Thinking about it in terms of how humans would approach it, one would need to describe specific characteristics of a particular sound to replicate it. Often this happens 'behind the scenes', as our perception does most of the work for us, and we never really explicitly think about all the separate trains of information that we have to process in order to achieve an understanding of sound.

In this specific case, we are approaching the programming of a granular synthesizer, therefore we need specific metrics of describing sound that can translate well to the parameters of a granular synthesizer.

2.1.2 Granulation

Automating parameters for granular synthesizer can be a difficult problem, considering how heavily the output sound depends on the audio being sampled. In other words the source of the grains.

To accommodate for that difficulty, some universal audio descriptors have to be used in order to describe the sound to the previously mentioned agent. These include (change when done) descriptions of rhythm, density, pitch etc, so that the input sound can be 'molded' into what will resemble the original sound that the agent is trying to replicate.

Concatenative synthesis - another thing widely used in speech synthesis offers a solution to this problem by creating a 'pool' of grains to choose from in order to replicate a sound (offer better description). It could be an interesting way to approach this problem, however a lot of research has been done in this domain (CataRT) and it seemed less experimental to tinker with. (the truth,

but obviously change later)

2.1.3 Machine Learning

A sophisticated neural network could be implemented to essentially build a granular synthesizer parameter space from the ground up using only the input sound, however implementing such a system and predicting its behaviour is a difficult task as well as a very experimental approach, that is quite difficult to control and predict.

Focusing on replicating certain aspects of the input sound seems like a more straightforward and most importantly predictable approach to this problem.

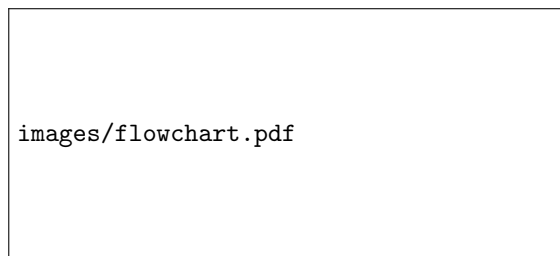
Investigating the correlation between certain audio descriptors and their influence over specific parameters of the synthesizer, and consequently separating the machine learning tasks to predict those aspects of the sound separately will result in a much more predictable outcome, as well as one that's easier to control.

Chapter 3

Methods

The method of tackling this problem will be based on previous research done in this domain. There are three essential components for this project:

Figure 3.1: Basic flowchart



The box “Granular Synthesiser” in figure 1. represents a granular synthesizer. It will be build in C++, using openFrameworks and the Maximillian library.

“Machine Learning” in figure 1, stands for Machine Learning. This aspect will most likely be implemented in python, either using the “scikit.learn” library, or the “TensorFlow” library. Depending on which Machine Learning algorithms will be used. This will be determined during testing. The ones being considered at the moment include k-NN, Neural Networks, and LSTM(Long Short-Term Memory Neural Network)

The “Audio Analysis” box in figure 1. represents the aspect of the project responsible for extracting audio features from input sounds. This requires some testing in order to see which works best for this task. Feature extractors considered are: MFCC and DBM. Environment choices for now include the python “pyAudioAnalysis” for Python, and “Essentia” library for C++.

Training of whichever algorithm chosen will be happening on data generated by me. Vectors of synthesis engine parameter values, and audio analysis values, possibly as a CSV file will be created. The goal here is to automate a sampled walk through the parameter space, and extract audio features for each point. This was suggested to me by Dr. Rebecca Fiebrink, during her office hours, as part of feedback for another assignment. Possibly, a genetic algorithm could be used for this process

Once all the pieces are working, the key thing is to make them all work together as soon as possible, to create a working prototype, from which, by user testing and iterative design I can move forward and improve the project tackling issues one by one.

The minimum viable prototype has to include every aspect from Figure 1., excluding any substantial testing. Meaning, that one audio feature extractor, one machine learning algorithm, and a synthesizer that work together would be enough. Any testing of possibilities, whether audio feature extractors, or machine learning algorithms is outside of the scope of a prototype. As long as each piece of software can communicate and produce a desirable end result, the prototype will be considered successful.

Chapter 4

Results and Analysis

Chapter 5

Discussion

5.1 Summary of Findings

5.2 Evaluation

5.3 Future Work

Start with a few sentences that summarize the most important results. The discussion section should be a brief essay in itself, answering the following questions and caveats:

What are the major patterns in the observations? (Refer to spatial and temporal variations.) What are the relationships, trends and generalizations among the results? What are the exceptions to these patterns or generalizations? What are the likely causes (mechanisms) underlying these patterns resulting predictions? Is there agreement or disagreement with previous work? Interpret results in terms of background laid out in the introduction - what is the relationship of the present results to the original question? What is the implication of the present results for other unanswered questions in earth sciences, ecology, environmental policy, etc....? Multiple hypotheses: There are usually several possible explanations for results. Be careful to consider all of these rather than simply pushing your favorite one. If you can eliminate all but one, that is great, but often that is not possible with the data in hand. In that case you should give even treatment to the remaining possibilities, and try to indicate ways in which future work may lead to their discrimination. Avoid bandwagons: A special case of the above. Avoid jumping a currently fashionable point of view unless your results really do strongly support them. What are the things we now know or

understand that we didn't know or understand before the present work? Include the evidence or line of reasoning supporting each interpretation. What is the significance of the present results: why should we care?

This section should be rich in references to similar work and background needed to interpret results. However, interpretation/discussion section(s) are often too long and verbose. Is there material that does not contribute to one of the elements listed above? If so, this may be material that you will want to consider deleting or moving. Break up the section into logical segments by using subheads.

Chapter 6

Conclusion

What is the strongest and most important statement that you can make from your observations? If you met the reader at a meeting six months from now, what do you want them to remember about your paper? Refer back to problem posed, and describe the conclusions that you reached from carrying out this investigation, summarize new observations, new interpretations, and new insights that have resulted from the present work. Include the broader implications of your results. Do not repeat word for word the abstract, introduction or discussion.