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in

Your Department

Your Advisor, Chair

First Committee

Second Committee

Third Committee

Last Committee

December 4, 2020

Blacksburg, Virginia

Keywords: Some Keywords, Subject matter, etc.

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ABSTRACT

Lorem ipsum dolor sit amet, consectetur adipiscing elit. Ut purus elit, vestibulum ut, placerat ac, adipiscing vitae, felis. Curabitur dictum gravida mauris. Nam arcu libero, nonummy eget, consectetur id, vulputate a, magna. Donec vehicula augue eu neque. Pellentesque habitant morbi tristique senectus et netus et malesuada fames ac turpis egestas. Mauris ut leo. Cras viverra metus rhoncus sem. Nulla et lectus vestibulum urna fringilla ultrices. Phasellus eu tellus sit amet tortor gravida placerat. Integer sapien est, iaculis in, pretium quis, viverra ac, nunc. Praesent eget sem vel leo ultrices bibendum. Aenean faucibus. Morbi dolor nulla, malesuada eu, pulvinar at, mollis ac, nulla. Curabitur auctor semper nulla. Donec varius orci eget risus. Duis nibh mi, congue eu, accumsan eleifend, sagittis quis, diam. Duis eget orci sit amet orci dignissim rutrum.

Dedicated to Virginia Tech.

Acknowledgments

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Chapter 1

Introduction

- Introduce the idea of musical research with computers. Talk about the illiac suite [?] and Music Information Retrieval.
- Significance of machine learning on the field
- Introduce idea of expressive musical performance. Brief conversation about the different performance components (articulation, dynamics, timing).
- Using Transformer architecture which hasn't been done in the field.
-

Richard: report results

Chapter 2

Background

There are two major research components that this project is based on. The first is the problem domain of expressive musical performance (EMP), and the second is the ML modeling domain of Transformers. We will introduce both of these components and provide context for what makes them interesting as a research project and why they are worth exploring together. We start first with an overview and definition of EMP, and then a summary of the Transformer.

2.1 Expressive Musical Performance

EMP is a subset of the research field of Music Information Retrieval (MIR) ¹ whose purpose is to use computational information to study, interpret, and gain a better understanding of the *essence* of music itself [?]. Perhaps the most well known MIR application is that of a musical recommendation system used by streaming services such as Spotify ² to provide a personalized and unique experience for each user. However, as Widmer [?] suggests, there are a number of other non-trivial problems that face the field and will require significant effort from the research community to properly understand. A proper understanding of musical performance is one of them.

¹Widmer[?] points out that MIR itself does not encompass the entire scope of computer music research, but that it is a good proxy to use when referring the field as a whole. We will operate under the same assumption

²[spotify.com](https://open.spotify.com/)

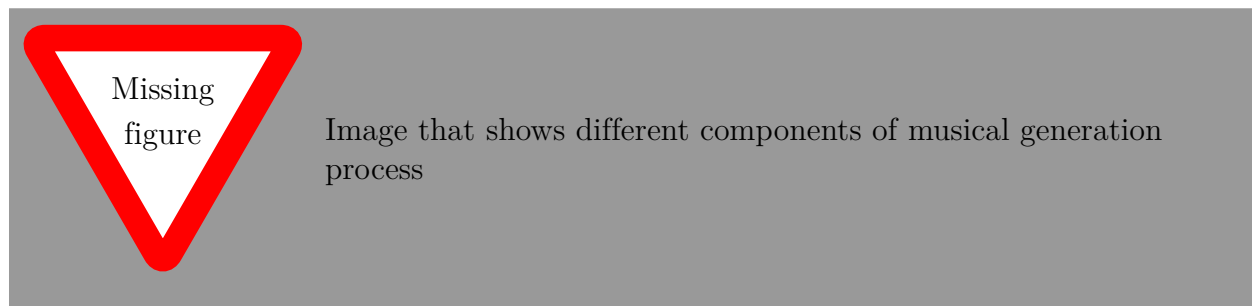


Figure 2.1: The first step of musical generation is composition, shown as a score in the figure. The second is performance, which is our area of interest. The third is the production of sound. Each different agent: composer, performer, instrument, and listener, can be thought of as a separate computational model in the generation process

MIR tasks can be broadly categorized in two ways - the first as computational methods for music analysis, and the second as computational methods for music generation. We are interested in the latter and its application in musical performance. In order to study how musical performance generation (and more particularly *expressive* musical performance generation) models work, it is necessary to gain a proper understanding of the entire computational musical generation process as a whole. ?] break the process down into 3 different components, with 4 different roles or agents that interact with that process. Figure 2.1 shows each step in the process as well as the agents that participate

Richard: Try to get permission to reproduce the image in the paper

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An EMP model is analogous to the performer as show in 2.1, who takes as input a musical composition and produces as output a performance. It is the phenomena of musical expression that makes the performance generation process interesting. Musical expression can be thought of as the performers' interpretation of a composition codified into different performance parameters that are intended to increase the quality of the musical experience by the final listener. Because the quality of a musical experience is highly subjective, there

is no definition of what makes for a "correct" interpretation of a given composition [?]. The subjective nature of EMP generation makes it a difficult problem to understand from a computational perspective. However, it also makes it a highly intriguing research topic given that a clear understanding of the problem from a computational perspective will no doubt further our understanding of what exactly it is that makes music so subjective in the first place, and bring us one step closer to understanding music itself.

To properly understand exactly what it is that constitutes expression in musical performance, it is necessary to provide a detailed description of the first two components of the generation process - namely, scores and performances. We refer the reader to appendix A.1 which provides some basic terminology and concepts that will be useful for grasping the following section ³. Due to the constraint of our data we focus only on western classical piano music.

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erence

2.1.1 Scores

A musical score is a symbolic representation of a musical composition. The symbolic notation used to create musical scores can be thought of as a language used to express musical ideas and information. It presents this information in a hierarchical structure with different levels of musical detail at each level. The lowest level contains information about the pitch and timing of every single note, as well as optional information about how the note should be played. This can include information specific to instruments such as the bow direction of a violin, but for our purposes (dealing only with piano) we will consider this to be the articulation of each note, usually indicated by legato or staccato

³Most of the appendix material may seem elementary to those who already have a background in music or musical notation. However, we feel that is necessary to include if for no other reason than to provide a clear definition for our descriptions both in general and at detailed mathematical level

Richard: Make sure to have some background information on articulation in the appendix

The middle level contains information related to certain substructures within the musical composition, which are usually expressed within a grouping of notes or measures. The most common score annotations at this level are dynamic markings which indicate whether to play a grouping of notes loud (Forte), soft (Piano), or to gradually increase or decrease the volume (crescendo or decrescendo). Although dynamic markings are the most common at this level, it is also possible to see score markings for all other musical features, such as local tempo or articulation of a certain substructure. Perhaps the most important score marking at this level is that of a phrase, which is a marking that indicates that a group of notes should be interpreted as belonging to a singular musical idea and that each note should fit within the context of the phrase as a whole. A phrase can be expressed through all of the different aforementioned musical features, including the tempo, timing, dynamics, and articulation of the notes.

The highest level contains meta-information that relates to the entire composition as a whole. This information typically includes the key signature and time signature, as well as the global tempo for the entire piece, most commonly represented as BPM.

2.1.2 Performance

An expressive musical performance contains most of the same musical information as does a score, but with one key difference; that is, that an expressive performance will deviate (or interpret) from the exact information that is presented in the score. For example, although a score may indicate a tempo of 120 BPM, it is highly unlikely that a given performer will perfectly adhere to this tempo throughout the entirety of the piece. This is even more

apparent if the score indicates a change in tempo somewhere in the composition. If a score indicates that the performance should speed up over a series of notes, there is no telling at what rate the tempo should increase. Some performers may choose to speed up at a fast rate and over a short period of time. Others may choose to increase the tempo at a slow rate and over a longer period of time. A single *accelerando* (a score indication to pick up the tempo) can result in either of these outcomes.

With that being said, a performance contains most of the same features related to a score, which include pitch, tempo, timing and articulation. Each of these expressive features will be measurable and absolute, whereas the score markings of these features can be viewed more as a suggestion than a rule. There are a few additional features that are present in performances which are not in scores. The first we will refer to as deviation which is heavily related to timing. It is typically represented as a numerical number which represents how far off the timing of a particular note deviates from its "correct" position in the score. These micro-timing deviations present in musical performances are an essential part of expression. Without them, indicating that each note onset and offset is exactly in line with its marking in the score, performances sound robotic and mundane.

The other important feature of performance that is not always present in a score applies specifically to the piano, and is the presence of a piano pedal. There are several different types of piano pedals, but the most common are the sustain pedal, which prolongs the duration of every note of the piano when activated, and the soft pedal which softens the sound of the entire piano. Although the effects of these pedals are directly related to the articulation and dynamics of the performance, their presence (or lack of) can be seen as a crucial component of piano performance. It is common for the sustain pedal to see active use in almost all modern piano performance, even when there doesn't exist any score marking indicating its use.

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Richard: Add section and reference to the specifics of feature engineering related to both the score and the performance in the methods section

2.1.3 Data

The data required for EMP generation includes some digital form of representation of a score as well as a corresponding performance. Scores are typically given in the form of MusicXML, which is a text-based representation of a score. Performances could be directly be rendered as audio which is the process used by human performers with the use of an acoustic instrument. Instead of audio however, an intermediate data form, MIDI, is used to represent the performance. This better aligns with the generation process outlined in [2.1](#). In the full generation process, a separate model would be used to take the performance data in MIDI and synthesize that into raw audio which would be presented to the listener. Both data formats contain all of the required information to represent all of the musical components of both a score and a performance, including pitch, tempo, timing, articulation, deviation, and pedal. See appendix [A.2](#) for more information on both MusicXML and MIDI.

To build an EMP generation model, it is necessary to run both the score and performance through a data alignment process in which every note of the performance is mapped to it's corresponding position in the score. Given the highly dynamic nature of musical performance, it is a non-trivial task to run this alignment process for a set of scores and performances, especially if the task is performed by manual human annotation. There exist methods for both manual and automatic alignment. Due to the time-consuming nature of manual alignment and the need for large data sets to build higher quality models, automatic alignment algorithms are an active area of research.

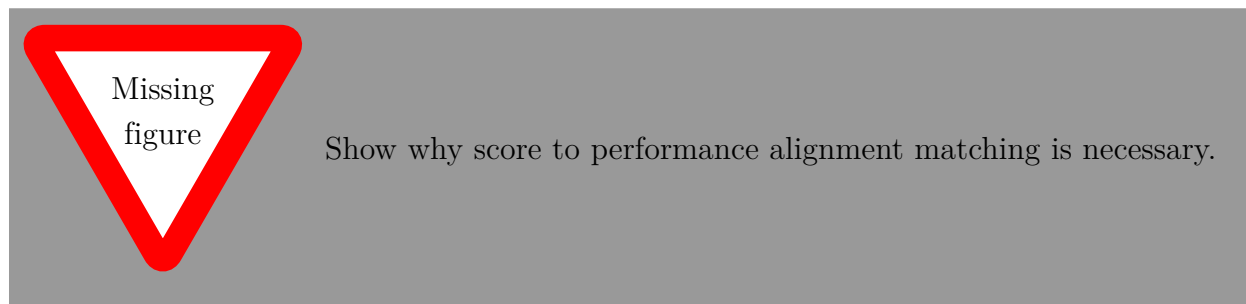


Figure 2.2: Two performances of the same score can vary wildly in their tempo and timing. This makes it necessary to have a score to performance alignment for every performance.

Richard: Add reference to section which gives relevant research

2.2 Transformers

To properly understand the significance of Transformers and their involvement in our work, it is necessary to provide context about the domain in which the Transformer was first introduced and give an overview of the existing work in that domain that the Transformer built on. We'll then provide some detail about the Transformer itself as well as adaptations of the original architectures and their results.

2.2.1 Natural Language Processing and Machine Translation

One of the most commonly studied fields in Machine Learning and Artificial Intelligence is Natural Language Process (NLP), which (similarly to MIR) uses computation to ascertain a better understanding of human language as well as build technological tools that are useful in performing common language tasks. One such task is that of machine translation, which involves using computational tools to translate text from one language to another. NLP research usually involves building sequence based models (which explore the individual

elements of an ordered set of items) due to the inherently sequential nature of language, as opposed to a non-sequential model which doesn't account for sequential data, such as a single image. Machine translation falls under the category of sequence-to-sequence (seq-2-seq) modeling problems, which involve the mapping and relationship of one sequence to another. This is typically in the form of translating a single sentence from one language (english) to another (french).

More specifically, machine translation (and other seq-2-seq tasks) can be defined as taking an input sequence $\mathbf{x} = \{x_1, x_2, x_3, \dots, x_m\}$ of size m and producing an output sequence $\mathbf{y} = \{y_1, y_2, y_3, \dots, y_n\}$ of size n such that $M(\mathbf{x}) = \mathbf{y}$, where M can be any machine translation model. In some seq-2-seq tasks, $m = n$ are the same, implying that the input and output sequence are the same length. As is often the case in language translation, the input sentence and output sentence are of varying lengths, so we can assume that $m \neq n$.

It is common to use an encoder-decoder architecture for M , where there is an encoder E which takes in the input data and outputs and finds some hidden representation $E(\mathbf{x} = \mathbf{z})$. This hidden representation is given as input to the decoder, and the decoder uses it to produce the final output, $D(\mathbf{z}) = \mathbf{y}$. We can then define an encoder-decoder seq-2-seq model as $M(x) = D(E(\mathbf{x})) = \mathbf{y}$. Historically, a Long-Short-Term-Memory neural network (LSTM)⁴ has been used for both E and D , where the hidden representation \mathbf{z} has been a fixed length vector .

One of the limitations of such a model is that it has to compress all of the information of the input data into the fixed length vector \mathbf{z} which causes the network to potentially lose important information, particularly in the case where an input sentence is given to the network which is longer than any present in the training data. [?] presents the attention

⁴An LSTM is a common variant of a Recurrent Neural Network (RNN) which is the most standard deep learning model used for sequence modeling. See https://en.wikipedia.org/wiki/Long_short-term_memory

mechanism which, used in conjunction with an RNN based encoder, allows for the hidden representation to itself be a sequence $\mathbf{z} = \{z_1, z_2, z_3, \dots, z_m\}$ of size m (the same size as the input sequence). Each z_i element in the sequence contains information about the whole input sequence, with an emphasis on the elements closest to the i -th element. This allows the hidden representation to encode any relationship that one element in the sequence has with another. The decoder then uses this information to "pay attention" to words in the output sequence that have a relationship with words in the input sequence, given the context that is encoded in the hidden representation at a particular time step i . The attention mechanism and model that uses it achieved state of the art results in the machine translation task, due in part to the fact that hidden representation is not limited to a fixed-size vector.

In his seminal paper, [?] introduces the Transformer architecture - an attention only sequence neural network that achieved new state of the art results in neural machine translation tasks. The Transformer came as an alternative Deep Learning modeling method to the Recurrent Neural Network (RNN) and its common adaption as a Long Short Term Memory (LSTM) network in sequence modeling problems, specifically in the domain of Natural Language Processing (NLP). There have been several adaptations of the original architecture proposed by [?] which have also achieved state of the art results in many NLP tasks, such as the General Language Understanding Evaluation (GLUE) benchmark [?][?], reading comprehension, question answering, and textual entailment [?] ⁵.

The original Transformer relies solely on the use of attention [?], more specifically self-attention, mechanisms to build an encoder-decoder ANN that was used for a neural machine translation task, which involves translating text (usually in the form of a single sentence) from one language to another using a single ANN model. Machine translation can be

⁵Some have called the Transformer and its applications the "ImageNet moment for NLP", in reference to the famous AlexNet Convolutional Neural Network architecture introduced in 2012 [?] that has spearheaded the rise in Deep Learning research ever since

Richard
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attention

seen as a standard seq-2-seq modeling problem, which involves mapping a input sequence $\mathbf{x} = \{x_1, x_2, x_3, \dots, x_m\}$ of size m to an output sequence $\mathbf{y} = \{y_1, y_2, y_3, \dots, y_n\}$ of size n . An encoder-decoder type model is typically used for a seq-2-seq problem. In this model, the encoder E will take the input sequence as input and generate a hidden encoding $E(\mathbf{x}) = \mathbf{z}$. The hidden encoding is fed as input to the decoder D which outputs the target sequence $D(\mathbf{z}) = \mathbf{y}$. Historically, RNN based encoder-decoder models would use a fixed-sized vector for the hidden representation \mathbf{z} . However, this fixed-length vector was known to have limitations, specifically with sentences longer than those present in the training data [?]. The introduction of attention allows for the hidden representation to itself be a sequence $\mathbf{z} = z_1, z_2, z_3, \dots, z_m$ of length of m . The decoder decides which parts of the hidden representation to "pay attention" to, and uses this context to decide which word should be the next output of the sequence [?]. This allows for information to propagate through the network without having to compress all of the information into a single fixed-sized vector. This mechanism achieved new state of the art results in machine translation .

The use of attention in sequence modeling (usually paired with an RNN) provided a way to create additional context and better memory across longer sequences, which had been a known limitation recurrent models . It wasn't until the introduction of the Transformer that the attention mechanism was used completely outside of any other existing modeling architecture, and formed the basis of a model.

The original Transformer [?] was also created for the neural machine translation task, but built both the encoder and decoder using a layered stack of self-attention (a form of attention which attends to only an input as opposed to using attention to find a relationship between an input and an output) mechanism along with a standard feed-forward neural network (FFNN). This attention only network architecture achieved new state of the art results in neural machine translation.

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2.3 Evaluation

- Evaluation is particularly difficult for a problem like EPG because there is no "correct" interpretation of a score. However, there is at least a vaguely understood relationship between a score marking and how a performer should use that marking within the context of a performance. For example, if a crescendo marking is used in a score, the performer should at the very least increase the volume of the performance relative to the current volume of the piece. The amount which the volume should increase or the rate at which it increases are not clearly defined, but the fact of the increase of volume itself is. This is the fundamental intuition behind the motivation to build computational models for expressive performance. Nonetheless, it still remains a difficult job to evaluate a given EPG model because of the ambiguity of what is "correct" or not.
- Evaluation methods used so far in EPG models are broken into two categories, quantitative and qualitative.
- Quantitative:
 - This follows standard techniques for experimentation of evaluation of ML models in general. It usually involves calculating a numerical value for a models inference on a separate test data set that was not used for model training or model selection.
 - Common metrics for regression like problems are mean squared error (MSE) and the pearson correlation coefficient (R2).
 - Due to the nature of EPG model evaluation mentioned above, it is not clear that "better" quantiative metric score for a given model over another indicates that the performance of the model is superior. .
- Qualitative

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- Qualitative evaluation methods involve gathering human feedback by playing performances of a given models performance to an audience and getting ratings or judgement of the model according to a predefined questionnaire or survey method. The nature of these evaluation methods is not consistent in the current literature and remains a challenge for the field to solve in the future. .

– Richard: Conduct more research for reference on current methods for qualitative evaluation

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Chapter 3

Related Work

3.1 Existing EPG models

- KTH system [?]. A rule-based system for expressive performance. Rules are selected through a empirical process based on human feedback.
- YQX. A Bayesian network that models timing, dynamics, and articulation [?]. Won the 2008 RenCon contest.
- Basis Function Models [?]
 - Linear Basis Functions. Uses Least Squares regression and Bayesian models with about the same performance
 - Non-Linear Basis Functions. Uses standard feed-forward network. FFNN perform better than Linear models. Also uses an RNN.

- Giraldo and Ramirez use several different ML algorithms, including Decision Trees, k-NN, SVM's, and FFNN to build an expressive performance generation system for improvisational Jazz guitar [?].
- Moulieras and Pachet use a Maximum Entropy model to infer the underlying distribution of expressive performance and build a generation system trained from a mix of popular music. Their expressive model outperforms base models in listening tests [?].

Richard

This
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Lot of
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- Jeong builds two versions of virtuosoNet, one using a recurrent hierarchical attention network (HAN) [?], and another using a recurrent graph network [?]. These models are built using a dataset order of magnitudes larger than other datasets and attempt to model the expressive performance feature of the pedal, which no other model does. The code for the models is also open source so it was chosen as the starting place for this work.

Richard: Add more papers and expand upon the existing research a bit more. Isn't completely necessary but will be good for my overall understanding

3.2 Datasets

- Talk about the fundamental limitations of gathering data for this problem, especially in relation to other fields [?]. Because of this, the lack of high-quality data is limited.
- The dataset used for the virtuosoNet [?] [?] will be the dataset used for the experiments. At the time the experiment started it was the largest publicly available dataset applicable to the EPG systems, and was chosen for use. A recent publication [?] builds off of the dataset used for the virtuosoNet with more sophisticated alignment and some extensions to the size (dataset is named ASAP). ASAP would be more appropriate for future use.
- One of the necessary data processing tasks for EMP is the alignment between the score and performance of a given piece. Because there is always an inherent interpretation of a composition by a performer Richard: Reference this in the introduction, there is no clear mapping between any given score and performance. It remains necessary to have some sort of alignment process to match each note in the performance with its related position in the score.

Richard: This needs more research. Find relevant papers to cite, as well as show a diagram that makes it clear why alignment is necessary

.

Chapter 4

Experiments

4.1 Model and Experiments

- Due to the open-source nature of virtuosoNet project and its attempt to build a more cohesive EPG model by introducing the pedal as an expressive feature and training on a much larger dataset, we built off of this model.
- Because of the significant advances in other sequence modeling domains (such as NLP) and the indication of increased performance of another related task with the Music Transformer [?], the main question we want to answer is whether we can see similar increases in model performance by applying a Transformer ANN architecture to the problem domain.
- We will experiment with a transformer encoder only architecture similar to BERT. The problem includes a 1-1 mapping between every note in the score and a related note in a performance. This is different than seq-2-seq modeling problem such as neural machine translation which maps a sequence of one length to another sequence of a different length, which is what the full Transformer architecture was intended for. The Transformer Encoder can be seen as a large encoder that learns the best representation for a given feature set. The model we'll build will use a simple FFNN that accepts the output of the transformer encoder to decode this representation and

give the final feature set which is then used to create a performance. This is similar to the BERT architecture and it's intended application.

Richard: Come up with a more detailed explanation of this modeling choice. Also create a visual diagram that explains the transformer encoder with the simple regression model sitting on top of it

- Because we are using the same dataset used to train virtuosoNet, we will directly compare the performance a Transformer model to the existing virtuosoNet models using the same quantitative metric, MSE.

Richard: Come up with specific model experiments and comparison in a table. Table doesn't have to have results but needs the general outline that will be used in the final paper

4.2 Evaluation

- Quantitative: Because we are using the same dataset used to train virtuosoNet, we will directly compare the performance a Transformer model to the existing virtuosoNet models using the same quantitative metric, MSE.

Richard: Come up with specific model experiments and comparison in a table. The table doesn't have to have results but needs the general outline that will be used in the final paper

- Due to time and resource constraints, no sophisticated qualitative evaluation was conducted for the models. However, a personal evaluation was used during the entire model development process.

Richard: Talk about method used for personal analysis

-

Chapter 5

Results

5.1 Quantitative

Add table with results of experiments along with explanations.

5.2 Qualitative

Give personal qualitative report.

Chapter 6

Discussion

Richard: The following are some interesting discussion ideas that have come up so far. There is no telling if these will be in the final paper or not after conducting more experiments.

- Transformer performs worse according the quantitative metrics. This could be because it doesn't build in a specific hierarchical layer that is specific the problem. It is a much more generic model. There is a lot of room for exploration into experimenting with different architectures based on the Transformer to better fit the problem domain.

Richard: Add more discussion based on more results

- Transformer appears to be a more dynamic model than the recurrent virtuosoNet model that makes more "mistakes". Does this mean that it is more "human".
- Pedal in performance is messy. Could be because of problems in the feature and modeling, or could just be because it is a difficult problem to model.
-

Richard: Discussion on qualitative results

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Appendices

Appendix A

Appendices I

A.1 Musical Concepts and Terminology

A.1.1 Pitch

The first and most basic component in music is pitch. Pitch is a perceptual property of sounds that relates to the physical frequency of a sound vibration [?]. It is what determines whether or not a sound can be thought of as "high" or low". The most commonly known way to conceptualize pitch is the 88 different keys on a piano keyboard, where each key represents a different pitch value. Pitch is most commonly labeled using scientific pitch notation, which couples a range of letters (A to G) with a range of numbers (zero to eight) that correspond to different octave ranges ¹. The most well known pitch is C4, or "middle C", and lays in the very center of a standard 88 key piano.

A.1.2 Tempo and Timing

Tempo in music describes the rate at which notes are played, and timing describes when a particular note should be played relative to the start of the composition. They are best explained in the context of modern western musical notation introduces the idea of note

¹https://en.wikipedia.org/wiki/Scientific_pitch_notation

durations, time signatures, measures, and beats ².

Richard: Find a more intuitive way to explain this. The piano roll explanation and visualization may work better

Each composition is broken down into a sequence of measures, and the time signature defines how many beat exist per measure, as well as the duration of a single beat. For example, a 4/4 time signature indicates that there are 4 beats per measure (the top half of the time signature), and that the duration of each beat is represented by a quarter note. A 3/4 time signature would indicate only 3 beats per measure, with the beat duration represented by a quarter note. The timing of a note would refer to it's measure, beat, and note duration. Tempo is most commonly given in beats per minute (BPM). A composition with a 4/4 signatue and a 120 BPM would mean that after one minute, 30 measures of the composition

should have been played so far.

A.1.3 Dynamics

Dynamics can simply be thought of as how loud or soft a note should be played (or has been played).

A.2 Data Representation

A.2.1 MusicXML

A.2.2 MIDI

²See https://en.wikipedia.org/wiki/Musical_notation#Modern_staff_notation for a more detailed explanation

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