

THIRD YEAR PROJECT REPORT

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Abstract

Faculty Name

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Third Year Project

3YP Title Here

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The Project Abstract is written here (and usually kept to just this page). The page is kept centered vertically so can expand into the blank space above the title too...

Contents

Abstract	ii
1 Project Definition	2
1.1 Introduction	2
1.1.1 Musical terminology	3
1.1.2 Subsection 2	3
1.2 Main Section 2	3
2 Product Design	4
2.1 Main Section 1	4
2.1.1 Subsection 1	4
2.1.2 Subsection 2	4
2.2 Main Section 2	5
3 Data	6
3.1 Main Section 1	6
3.1.1 Subsection 1	6
3.1.2 Subsection 2	6
3.2 Main Section 2	7
4 Model	8
4.1 Introduction	8
4.2 Model Requirements	8
4.2.1 MVP Requirements	8
4.2.2 Other possible Requirements	9
4.2.3 Evaluation criteria	10

4.3	Models	11
4.3.1	Model evaluation template	12
4.3.2	HMMs	12
	Explanation and Applicability	12
	Implementations	13
4.3.3	DNN-HMMs	14
	Explanation and Applicability	14
	Implementations	14
4.3.4	GANs	14
	Explanation and Applicability	14
	Implementations	15
4.3.5	RNNs	15
4.3.6	LSTMs	16
	Implementations	17
4.3.7	Transformers	17
4.4	Our Model	18
4.4.1	Model Functionality	18
4.5	Training	20
4.5.1	Avoiding Overfitting	20
4.5.2	Issues	20
4.5.3	Testing	21
4.6	Results	21
5	Voice Processing	22
5.1	Main Section 1	22
5.1.1	Subsection 1	22
5.1.2	Subsection 2	22
5.2	Main Section 2	23
6	Chapter Title Here	24
6.1	Welcome and Thank You	24

6.2	Learning L ^A T _E X	24
6.2.1	A (not so short) Introduction to L ^A T _E X	25
6.2.2	A Short Math Guide for L ^A T _E X	25
6.2.3	Common L ^A T _E X Math Symbols	25
6.2.4	L ^A T _E X on a Mac	25
6.3	Getting Started with this Template	26
6.3.1	About this Template	26
6.4	What this Template Includes	27
6.4.1	Folders	27
6.4.2	Files	27
6.5	Filling in Your Information in the <code>main.tex</code> File	29
6.6	The <code>main.tex</code> File Explained	29
6.7	Thesis Features and Conventions	31
6.7.1	Printing Format	31
6.7.2	Using US Letter Paper	31
6.7.3	References	31
	A Note on bibtex	32
6.7.4	Tables	33
6.7.5	Figures	33
6.7.6	Typesetting mathematics	35
6.8	Sectioning and Subsectioning	36
6.9	In Closing	36
A	Frequently Asked Questions	37
A.1	How do I change the colors of links?	37
	Bibliography	38

List of Figures

4.1	An RNN	15
4.2	The repeating module in an LSTM from Oinkina, 2015	16
4.3	The architecture of a transformer from Vaswani et al., 2017	18
6.1	An Electron	34

List of Tables

4.1	The parameters of the command line interface for the model	19
6.1	The effects of treatments X and Y on the four groups studied.	33

Chapter 1

Project Definition

1.1 Introduction

The process of writing music often begins with an idea for a melody which is followed by the development of chords to accompany it. The matching of chords to a melody is a skill which usually requires years of training in musical techniques such as harmonisation. There is a large market of amateur musicians who lack the necessary training for this task but would otherwise enjoy the experience of developing music. In this report we will detail the design of **INSERT NAME HERE**, a system which facilitates the generation of an appropriately matched set of chords to a given monophonic melody. Users are able to record a melody using their microphone and regenerate the chord sequence until they feel they have found one suitable for the intended feel of their song. Songs and generated chord accompaniments can be played back to the user, saved to a library of songs and shared using our songwriting community feature.

The problem of converting a recorded melody to a set of chords can be decomposed into a set of simpler sub-problems, these being: The conversion of the recorded melody into a form which numerically represents its features. The generation of a set of chords from this numerical representation. The latter of these two problems can be solved in many ways many of which require a detailed knowledge of music to find patterns in melodies to match to chords. **WHY IS THIS A PROBLEM FOR US** A method which requires little knowledge of music is the use of a machine learning model to extract these patterns from data of known chord melody pairings from professionally composed music. The curation and processing of an appropriate dataset requires significant effort and care in itself. This leads to a natural division of labour across the project into the categories of: User Interface and general product design - detailed in [2](#) by Di Wan; Curation and

preparation of a dataset in an appropriate format - detailed in 3 by Terence Tan; The design of an appropriate machine learning model - detailed in 4 by Edward Gunn; The conversion of recorded melody to the same format as expressed in the dataset - detailed in 5 by Kitty Fung.

1.1.1 Musical terminology

Through this report we will use a variety of musical terminology which will be defined here. A piece of music is composed of a sequence of adjacent **measures**, periods of time in which notes and chords can be played. We will generally take a **measure** to mean a bar in the music, however it is not restricted to this. We restrict **chord** to refer to a triad of notes, the justification for this is explained in **reference to explanation**. The use of **melody** refers to a monophonic melody in which only one note is played at a time, excluding accompanying chords, unless otherwise stated.

1.1.2 Subsection 2

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Chapter 2

Product Design

2.1 Main Section 1

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Chapter 3

Data

3.1 Main Section 1

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Chapter 4

Model

4.1 Introduction

The problem of generation of a set of chords from a melody is very similar to that of translating one language to another. The translation problem is one that is very popular in machine learning research and thus there is many resources on it. However the music related problem is harder to solve due to the extra dimension each of its elements contains. Each element in a melody has both pitch, represented by a discrete symbol or note, and duration whereas each element in the sequence of language is composed of only the discrete symbols or words. Therefore in order to use techniques developed for natural language processing it is necessary to encode the melody and chords in such a way that their dimensions are collapsed into one. Since this collapsing of dimensions has already been conducted in **REF TO DATA CHAPTER** we are able to take full advantages of NLP techniques. There has been many attempts to apply some of these techniques to the chord generation problem. In order to find the best model we evaluate these attempts against a defined criteria and develop a novel model to evaluate against the same criteria.

4.2 Model Requirements

4.2.1 MVP Requirements

The MVP requirements were defined at the beginning of the project to ensure it is compatible with the other parts of the product that were simultaneously in development and to ensure the possibility of the creation of the prototype for proof of concept. We required the model to be a black box in which we could input a melody and it would output chords which sounded good. The notion of

sounding good is intentionally vague as what sounds good or bad is usually down to individual taste in music. There is never a definitive answer to which chords would sound the best. However, we felt **NEED STRONGER PROOF** that even people without any musical training would be able to judge whether chords fit the melody relatively accurately. Within this overarching requirement we defined tighter constraints for the sake of both practicality and user experience. The model would have to be designed for sequential data such that the temporal relationship of the different notes being played were taken into account. It would be possible to simply learn a function where for a given measure a chord suited to the notes played within that measure was generated without taking into account surrounding measures. This approach could produce reasonable results and be significantly more simple than other options, however, the quality and variation of chords generated would be lower than that of a model for sequential data, hence our choice to forego it. The model would also have to be conditional. For a given input it should produce a catered output. This is an obvious constraint however for models such as a GAN it requires changes to the format of the input data. To maximise user experience the models should be non-deterministic. This allows users to regenerate the accompaniment multiple times to obtain new chords and thus means they can choose what they feel is best. For practicality sake we constrained our MVP to only require one chord to be played each measure as the problem of determining when a chord should be played is of similar difficulty to choosing which chord to play.

4.2.2 Other possible Requirements

There were some constraints which were not used in our project which would provide better quality chords but provided too large a practical barrier to be implemented. The act of collapsing the pitch and duration of the notes into a single dimension removes information from the melody. In our case this information is the order in which the notes are played and the rhythmic intention of the composer. It is likely a model would be able to produce more suitable chords given this information. Thus, for a stricter set of constraints we could include the requirement that the order and duration of notes are both maintained in the data. The accompaniment for music is not limited to a single chord played at the start of each bar. A much more interesting accompaniment would be generated if it were not limited to this restrictive format. As shown in Zeng and Lau, 2021 it is possible to create a model which learns the divisions within the music and thus learns when would be most

appropriate to play a chord. Therefore, in order to allow for more interesting accompaniment, the requirement of one chord per bar could be changed such that chords should be played with closer freedom to that of a human composer.

4.2.3 Evaluation criteria

Approach Many attempts to solve the chord generation problem have been made each of which containing variants on models and implementations which result in large variations in metrics used for evaluation. There is also a large variation in features for which quantitative evaluation is impossible. As it would be impractical to implement each relevant model ourselves to allow for full standardisation of evaluation we will evaluate models built ourselves and from previous work based on a set of criteria defined below.

Data One of the biggest influences on the effectiveness of a model is the dataset on which it is trained. In general a larger dataset results in a better generalisation of learning **Need proof**. Most datasets are comprised of lead sheets which can be translated into chord melody pairs. Many datasets are then further divided down into measures in which there is one chord played per measure and the melody within that measure is somehow encoded. Thus for best comparison the number of measures used in the dataset is a good criterion for evaluation. This is a particularly important area for evaluation as it will strongly affect the output of the model and thus the results of other

Quantitative Evaluation There is difficulty in quantitative evaluation for this problem as there is no strictly correct chords for a given melody and thus comparison to any specific set of chords gives a skewed interpretation of the output. However, the use of conventional quantitative evaluation does still correlate strongly with the sentiment given by a human judging the quality of chords and thus can be cautiously used in evaluation. The test set from the data can be used to compare the outputs from the model to previously assigned chords. Accuracy can be found by finding the number of chords successfully predicted and dividing by the total number of chords produced.

$$\text{Accuracy} = \frac{\text{Number of Chords Correctly Assigned}}{\text{Total Number of Chords Generated}} \quad (4.1)$$

As this was the only quantative measure consistently used across previous implementations this is the only one we will consider for evaluation.

Qualitative Evaluation Some previous work Simon, Morris, and Basu, 2008, Lim, Rhyu, and Lee, 2017 carried out experiments to judge the sentiment of untrained musicians to the generated chords. Participants were played the melody accompanied by a varying set of chords and asked to judge which chords they felt were best out of a set containing chords generated by models and some human written accompaniment. The results from these experiments can be used to compare models to each other as well as evaluate them relative to the standard set by human written chords. As well as evaluation based on the quality of chords produced we will discuss extra functionality made possible by some models and the effect this has on other evaluation metrics.

Criteria The final criteria used for evaluation are thus:

- Number of measures in the dataset
- Accuracy in testing
- Human sentiment towards generated chords
- Extra functionality the model provides

4.3 Models

The generation of chords to accompany a monophonic melody is an area which has developed as machine learning techniques are improved. An early approach to this problem from CUNHA and RAMALHO, 1999 used a standard MLP to learn the relationship between melodies and accompanying chords. MySong, the first attempt focusing specifically on a vocal melody in Simon, Morris, and Basu, 2008 used an augmented hidden markov model with parameters such that users could adjust the "Happy factor" and "Jazz factor" to alter the mood of the generated chords to their preference. A weakness of the hidden markov model is that at each timestep only the previous timestep is considered when suggesting a chord. Lim, Rhyu, and Lee, 2017 utilises a BLSTM to increase the effect of long term dependencies on the output rather than relying only on the previous output. Chen and Qi, 2015 tested and evaluated a number of more simple models such as Logistic Regression, Naive

Bayes, SVM and Random Forest trained on data from 43 lead sheets

Zeng and Lau, 2021 proposed a model that learns a structured representation for use in symbolic melody harmonization. It utilises two layers of LSTMs to determine when each chord should be played then utilise a policy gradient RL method to select each chord. MuseGAN

ChordGAN

CLSTMGAN for melody Generation

4.3.1 Model evaluation template

Model explanation/history - why is it generally applicable

Previous implementations

General background

Evaluation

Data

Quantitative

Qualitative

Extra features

4.3.2 HMMs

Explanation and Applicability

Hidden Markov Models or HMMs were first put forward by Leonard E. Baum in a series of statistical papers **list papers from wikipedia**. They model a stochastic process in which the desired states, \mathbf{X} , are not directly observable but related states, \mathbf{Y} , which directly influence the desired states are. By modeling the \mathbf{X} and \mathbf{Y} Markov Processes **references?** we are able to infer the hidden state. To apply an HMM to our problem we take the hidden state, \mathbf{X} , to be the chords played, and the observable state \mathbf{Y} to be the Melody. It is notable that the conditional probability distribution of the hidden variable $x(t)$ at time t , only depends on that of the previous time step $x(t-1)$ and that $y(t)$ only depends on the hidden distribution at the current time step $x(t)$. For our purposes this

limits the "memory" the model could have to a single measure and thus seriously reduces the effect of relationship between chords across time.

Implementations

MySong The first application of an HMM to this problem while also being the the first attempt focusing specifically on a vocal melody from Simon, Morris, and Basu, 2008 used an augmented hidden markov model with parameters such that users could adjust the "Happy factor" and "Jazz factor" to alter the mood of the generated chords to their preference. No measure was given for the accuracy of the model, however a study which compared the quality of MySong to Manually assigned chords was carried out. In 264 comparisons the participants preferred MySon 95 times, manual 121 times and had no preference 48 times The additional user input possible with the "Happy factor" and "Jazz factor" parameters are reported to significantly improve the user experience, however, this is thier only implementation and so nothing can be infered about the quality of the HMM itself.

Chord Generation from Symbolic Melody As a comparison for the main model in Lim, Rhyu, and Lee, 2017 an HMM was implemented. The HMM is trained on a reduced version of the **Include link?** wikifonia.org dataset which includes and array of Western music genres with 2,252 lead sheets, 1802 of which are used for training. This overall comes to 72,418 measures for the training set and 17,768 for the test set. They tested the accuracy of the HMM on sequences of length 4, 8, 12, and 16 bars long gaining an accuracy of 0.4033, 0.4043, 0.4041, and 0.4045 respectively and an average accuracy of 0.4041. They also carried out a user subjective test with 25 participants each evaluating 18 sets, each set containing a melody accompanied by three generated chord sequences and an original chord sequence. The users would listen to the music played and judge the accompaniment on a scale of 1, being not appropriate, and 5, being very appropriate. The HMM achieved an average score of 2.31 and the original chords achieved an average score of 4.04.

Machine Learning in Automatic Music Chords Generation An HMM was tested in Chen and Qi, 2015 along with other simpler models. It was trained on 43 lead sheets, with 813 measures total. An accuracy of 0.4844 was achieved, however an choice of only 7 chords were used unlike the usual 24.

4.3.3 DNN-HMMs

Explanation and Applicability

A deep neural network HMM makes it possible to assume a posterior for the HMM using the output from the softmax output layer of the DNN. This allows for the posterior to be learned in training.

Implementations

Chord Generation from Symbolic Melody This was implemented much like the HMM from Lim, Rhyu, and Lee, 2017 mentioned above in 4.3.2. The same dataset was used and an accuracy of 0.4502, 0.4482, 0.4495, and 0.4468 was obtained on the 4, 8, 12 and 16 bars respectively. This results in an average accuracy of 0.4487. On the user subjective test the DNN-HMM achieved a score of 2.48.

4.3.4 GANs

Explanation and Applicability

Generative Adversarial Networks or GANs were first proposed in the landmark paper Goodfellow et al., 2014. Unlike most models GANs consist of two separate agents working against each other. The first model, the Generator, takes an input of random noise and outputs data which mimics the training data. The second model, the Discriminator, takes as input an example from the training data or an output from the Generator and outputs a value between 0 and 1 indicating whether it thinks the input is real or generated. The Binary Cross Entropy loss is used for the Discriminator averaged across real and generate examples. The loss for the Generator is the log complement of that for the Discriminator. Thus the two models play the following minmax game:

$$\min_D \max_G V(D, G) = \mathbf{E}_{x \sim p_{data}(\mathbf{x})} [\log D(\mathbf{x})] + \mathbf{E}_{z \sim p_{data}(\mathbf{z})} [\log(1 - D(G(\mathbf{z})))] \quad (4.2)$$

Theoretically the Generator and Discriminator can be any differentiable function thus leaving much room for flexibility. The problem with GANs for our uses is that they are not conditional, the only input to the Generator is noise. A proposed remedy for this was presented in Mirza and Osindero, 2014. By concatenating the label data, in our case the melody, with the noise as input to the Generator and doing the same with the training data and the label data, in our case a chord and

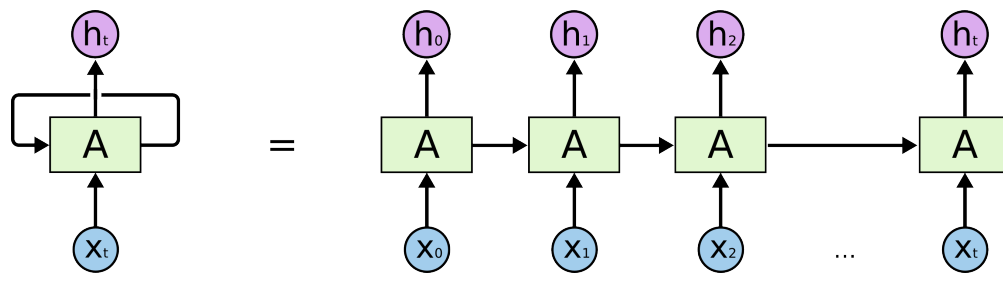


FIGURE 4.1: A many inputs to many outputs diagram of an RNN from Oinkina, 2015

melody, as input to the Discriminator GANs can be made conditional. Thus they would be suitable for our uses.

Implementations

ChordGAN Lu and Dubnov, 2021 uses a conditional GAN along with Chroma Feature Extraction to generate chords for a specific genre of music based on the dataset it was trained on. They used three different data sets one for each Pop, Jazz and Classical music each shorter than 1600 measures (specific lengths are not given). The model achieved an accuracy of 0.68, 0.74 and 0.64 respectively across the datasets.

4.3.5 RNNs

Recurrent Neural Networks or RNNs have become a staple in the machine learning engineer's library of models. They are structured much like a normal MLP, however, they contain an extra connection to the state of the network in the timestep before. This means that the state of the network at each timestep depends that of the previous timestep and thus the state at the current timestep is affected by the state in all previous timesteps. This makes RNNs ideal for processing sequential data as temporal relationships are taken into account. The gradient of RNNs can be found using a variation of the backpropagation algorithm usually known as backpropagation through time. RNNs that operate on large sequences often experience the problems of exploding or vanishing gradients leading to a saturation of learning.

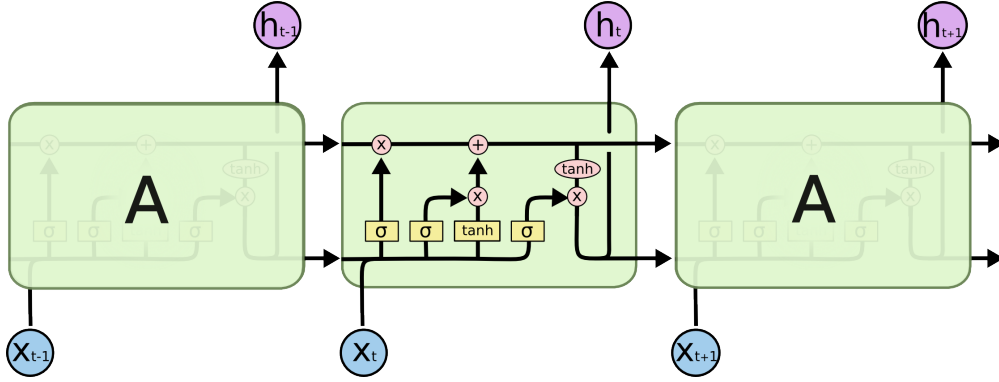


FIGURE 4.2: The repeating module in an LSTM from Oinkina, 2015

4.3.6 LSTMs

The Long Short Term Memory model or LSTM was proposed in Hochreiter and Schmidhuber, 1997 in order to overcome the gradient problems related to RNNs and thus allow for faster training on long sequences. They are also capable of learning longer dependencies due to their internal memory. An LSTM cell has three gates: input, forget, and output. The state of these gates determines whether the cell allows new input, forgets old information, and affects the output at the current timestep. At timeset t , the states of the gates are given by:

$$i_t = \sigma(w_i[h_{t-1}, x_t] + b_i) \quad (4.3)$$

$$f_t = \sigma(w_f[h_{t-1}, x_t] + b_f) \quad (4.4)$$

$$o_t = \sigma(w_o[h_{t-1}, x_t] + b_o) \quad (4.5)$$

where i_t , f_t and o_t denote the input, forget, and output gates state respectively, h_{t-1} is the output at the previous timestep. w and b represent weights and biases of each gate, x_t is the input to the LSTM cell, and $\sigma(\cdot)$ is the sigmoid function applied elementwise. The current output of the cell is computed by:

$$h_t = o_t \circ \tanh(c_t) \quad (4.6)$$

$$c_t = f_t \circ c_{t-1} + i_t \circ \tilde{c}_t \quad (4.7)$$

$$\tilde{c}_t = \tanh(w_c[h_t, x_t] + b_c) \quad (4.8)$$

Implementations

Chord Generation from Symbolic Melody This was the main model under scrutiny in Lim, Rhyu, and Lee, 2017 mentioned above in 4.3.2 and 4.3.3. They build a bidirection LSTM with two hidden layers and used a hyperbolic tangent activation function. The softmax function is applied to the output in order to represent the probability that each of the 24 chords is played. The same dataset as the HMM and DNN-HMM was used and an accuracy of 0.5055, 0.5032, 0.4923, and 0.4990 was obtained on the 4, 8, 12 and 16 bars respectively. This results in an average accuracy of 0.5000. On the user subjective test the DNN-HMM achieved a score of 3.55.

Automatic Melody Harmonization A particularly interesting model heavily relying on LSTMs surrounded by a reinforcement learning framework was proposed in Zeng and Lau, 2021. The simultaneously trained a Structured Representation Module responsible for learning note-level, phrase-level and segment level representations, a Segmentation Module acting as a reinforcement learning agent to decide whether the current note is the boundary of a phrase or segment and a Harmonisation Module responsible for generating chords for each segment. They used the Hooktheory Lead Sheet Dataset with 10,000 songs, no number of measures is given but with the difference in model architecture so drastic a comparison on this metric would be inappropriate anyway. They achieved an accuracy of 0.3742 and compared that to an SVM, CNN, LSTM and BLSTM with blocked Gibbs sampling which achieved an accuracy of 0.2516, 0.2664, 0.2802, 0.2933 respectively.

4.3.7 Transformers

Transformers have been a revolutionary development in the field of NLP, first proposed by Vaswani et al., 2017 they have become very widely used and regularly produce state of the art performance. Transformers utilise the encoder decoder model for sequence to sequence translation. They also heavily use attention which is a mechanism for weighting the importance of different elements of a sequence of data. Specifically they propose the Scaled dot-product attention shown in 4.9

$$\text{Attention}(Q, K, V) = \text{softmax}\left(\frac{QK^T}{\sqrt{d_k}}\right)V \quad (4.9)$$

where Q , K and V represent the queries, keys and values respectively and d_k is the dimension of queries and keys.

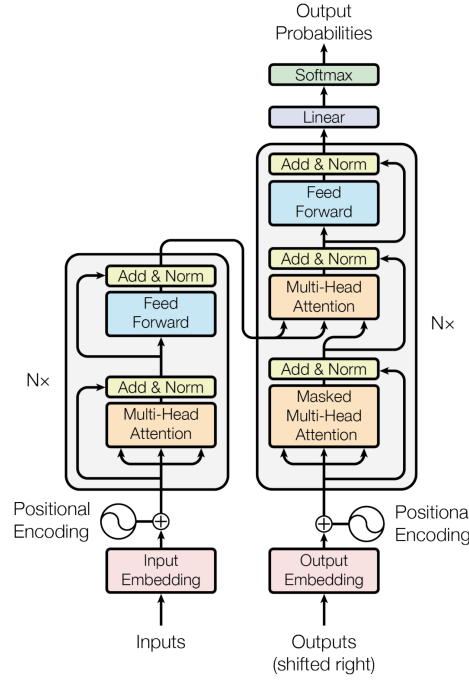


FIGURE 4.3: The architecture of a transformer from Vaswani et al., 2017

4.4 Our Model

We propose the use of a conditional BLSTM GAN to learn the association between melodies and chords. The Discriminator consists of a number of LSTM layers followed by a linear output layer. The Generator consists of a linear layer followed by a number of LSTM layers and then by another linear layer. The Binary Cross Entropy loss is used to determine the Discriminator loss for each example. This is averaged across every measure and then between real and generated examples

4.4.1 Model Functionality

As proof of concept for the design we developed a command line based interface for the model such that it was easy to train, vary parameters and test. In production the user would be able utilise a subset of this interface, such as the option to generate accompaniment and have it played back, through a graphical user interface. The options available in the interface are show in table 4.1

TABLE 4.1: The parameters of the command line interface for the model

Parameter	Options	Default	Description
-input_size	[input_size]	12	The size of the input vector to the discriminator representing the melody
-output_size	[output_size]	25	The size of the output vector representing the chord played in each measure
-h_size	[h_size]	128	The size of the hidden layers in the LSTM layers
-n_layers	[n_layers]	2	The number of LSTM layers in the generator and discriminator
-noise_size	[noise_size]	12	The size of the noise vector concatenated with the input vector to the generator
-max_seqlen	[max_seqlen]	200	The maximum length of song in measures used in training
-src_data	[src_data]		The default path to the training data
-batch_size	[batch_size]	10	The size of the batches used in the stochastic gradient descent algorithm
-epochs	[epochs]	100	The number of epochs used in training
-prntevery	[prntevery]	10	The number of epochs between printing an example during training
-load		False	Whether to load a model in or not
-load_dir	[load_dir]		The path to the folder containing pretrained models
-model_num	[model_num]	1	The number of the model to be loaded in
-save		False	Whether to save the model after training
-save_dir	[save_dir]		The path to the save directory
-playback		False	Whether to play an example of generated music at the end of training

4.5 Training

Optimiser We used the Adam optimiser Kingma and Ba, 2014 which is recommended in the Stanford CS231n: Convolutional Neural Networks for Visual Recognition¹ as the default optimiser.

EXPLAIN ADAM WHEN NOT FALLING ASLEEP

Loss As we are using a GAN we use the Binary Cross Entropy or BCE as our loss function

$$\frac{1}{N} \sum_{i=1}^N \log(p(y_i)) \quad (4.10)$$

which for the discriminator takes the form

$$\frac{1}{N} \sum_{i=1}^N \log(D(\text{conc}(x_i, z_i))) + \log(1 - D(\text{conc}(G(z_i), z_i))) \quad (4.11)$$

where x_i is are the real examples, z_i are the melodies and the *conc* function concatenates the two vector arguments. The loss is the sum of the loss of the discriminator on real data and generated data. The generator loss is

$$- \frac{1}{N} \sum_{i=1}^N \log(1 - D(\text{conc}(G(z_i), z_i))) \quad (4.12)$$

which results in trying to maximise the loss of the discriminator on generated data. This is equivalent to finding the discriminator loss on generated data when labeled as real which is the method we use in our implementation.

4.5.1 Avoiding Overfitting

Dropout GAN techniques Reducing number of LSTM layers

4.5.2 Issues

Training is conducted in batches of a given size. It is unlikely that the size of the dataset is divisible by the batch size so the final batch is likely to be shorter than the others. This initially caused problems, however, we were able to set the dimensions of relevant objects, such as the input noise to

¹<https://cs231n.github.io/>

the generator, to be relative to the current batch size rather than the usual batch size thus solving the problem. Softmaxing of outputs of generator Device management

4.5.3 Testing

4.6 Results

Chapter 5

Voice Processing

5.1 Main Section 1

Lorem ipsum dolor sit amet, consectetur adipiscing elit. Aliquam ultricies lacinia euismod. Nam tempus risus in dolor rhoncus in interdum enim tincidunt. Donec vel nunc neque. In condimentum ullamcorper quam non consequat. Fusce sagittis tempor feugiat. Fusce magna erat, molestie eu convallis ut, tempus sed arcu. Quisque molestie, ante a tincidunt ullamcorper, sapien enim dignissim lacus, in semper nibh erat lobortis purus. Integer dapibus ligula ac risus convallis pellentesque.

5.1.1 Subsection 1

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5.1.2 Subsection 2

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5.2 Main Section 2

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Chapter 6

Chapter Title Here

6.1 Welcome and Thank You

Welcome to this L^AT_EX Thesis Template, a beautiful and easy to use template for writing a thesis using the L^AT_EX typesetting system.

If you are writing a thesis (or will be in the future) and its subject is technical or mathematical (though it doesn't have to be), then creating it in L^AT_EX is highly recommended as a way to make sure you can just get down to the essential writing without having to worry over formatting or wasting time arguing with your word processor.

L^AT_EX is easily able to professionally typeset documents that run to hundreds or thousands of pages long. With simple mark-up commands, it automatically sets out the table of contents, margins, page headers and footers and keeps the formatting consistent and beautiful. One of its main strengths is the way it can easily typeset mathematics, even *heavy* mathematics. Even if those equations are the most horribly twisted and most difficult mathematical problems that can only be solved on a super-computer, you can at least count on L^AT_EX to make them look stunning.

6.2 Learning L^AT_EX

L^AT_EX is not a WYSIWYG (What You See is What You Get) program, unlike word processors such as Microsoft Word or Apple's Pages. Instead, a document written for L^AT_EX is actually a simple, plain text file that contains *no formatting*. You tell L^AT_EX how you want the formatting in the finished document by writing in simple commands amongst the text, for example, if I want to use *italic text for emphasis*, I write the `\emph{text}` command and put the text I want in italics in between the curly braces. This means that L^AT_EX is a "mark-up" language, very much like HTML.

6.2.1 A (not so short) Introduction to L^AT_EX

If you are new to L^AT_EX, there is a very good eBook – freely available online as a PDF file – called, “The Not So Short Introduction to L^AT_EX”. The book’s title is typically shortened to just *lshort*. You can download the latest version (as it is occasionally updated) from here: <http://www.ctan.org/tex-archive/info/lshort/english/lshort.pdf>

It is also available in several other languages. Find yours from the list on this page: <http://www.ctan.org/tex-archive/info/lshort/>

It is recommended to take a little time out to learn how to use L^AT_EX by creating several, small ‘test’ documents, or having a close look at several templates on:

<http://www.LaTeXTemplates.com>

Making the effort now means you’re not stuck learning the system when what you *really* need to be doing is writing your thesis.

6.2.2 A Short Math Guide for L^AT_EX

If you are writing a technical or mathematical thesis, then you may want to read the document by the AMS (American Mathematical Society) called, “A Short Math Guide for L^AT_EX”. It can be found online here: <http://www.ams.org/tex/amslatex.html> under the “Additional Documentation” section towards the bottom of the page.

6.2.3 Common L^AT_EX Math Symbols

There are a multitude of mathematical symbols available for L^AT_EX and it would take a great effort to learn the commands for them all. The most common ones you are likely to use are shown on this page: <http://www.sunilpatel.co.uk/latex-type/latex-math-symbols/>

You can use this page as a reference or crib sheet, the symbols are rendered as large, high quality images so you can quickly find the L^AT_EX command for the symbol you need.

6.2.4 L^AT_EX on a Mac

The L^AT_EX distribution is available for many systems including Windows, Linux and Mac OS X. The package for OS X is called MacTeX and it contains all the applications you need – bundled together and pre-customized – for a fully working L^AT_EX environment and work flow.

MacTeX includes a custom dedicated L^AT_EX editor called TeXShop for writing your ‘.tex’ files and BibDesk: a program to manage your references and create your bibliography section just as easily as managing songs and creating playlists in iTunes.

6.3 Getting Started with this Template

If you are familiar with L^AT_EX, then you should explore the directory structure of the template and then proceed to place your own information into the *THESIS INFORMATION* block of the `main.tex` file. You can then modify the rest of this file to your unique specifications based on your degree/university. Section 6.5 on page 29 will help you do this. Make sure you also read section 6.7 about thesis conventions to get the most out of this template.

If you are new to L^AT_EX it is recommended that you carry on reading through the rest of the information in this document.

Before you begin using this template you should ensure that its style complies with the thesis style guidelines imposed by your institution. In most cases this template style and layout will be suitable. If it is not, it may only require a small change to bring the template in line with your institution’s recommendations. These modifications will need to be done on the `MastersDoctoralThesis.cls` file.

6.3.1 About this Template

This L^AT_EX Thesis Template is originally based and created around a L^AT_EX style file created by Steve R. Gunn from the University of Southampton (UK), department of Electronics and Computer Science. You can find his original thesis style file at his site, here: <http://www.ecs.soton.ac.uk/~srg/softwaretools/document/templates/>

Steve’s `ecsthesis.cls` was then taken by Sunil Patel who modified it by creating a skeleton framework and folder structure to place the thesis files in. The resulting template can be found on Sunil’s site here: <http://www.sunilpatel.co.uk/thesis-template>

Sunil’s template was made available through <http://www.LaTeXTemplates.com> where it was modified many times based on user requests and questions. Version 2.0 and onwards of this template represents a major modification to Sunil’s template and is, in fact, hardly recognisable. The work to make version 2.0 possible was carried out by Vel and Johannes Böttcher.

6.4 What this Template Includes

6.4.1 Folders

This template comes as a single zip file that expands out to several files and folders. The folder names are mostly self-explanatory:

Appendices – this is the folder where you put the appendices. Each appendix should go into its own separate `.tex` file. An example and template are included in the directory.

Chapters – this is the folder where you put the thesis chapters. A thesis usually has about six chapters, though there is no hard rule on this. Each chapter should go in its own separate `.tex` file and they can be split as:

- Chapter 1: Introduction to the thesis topic
- Chapter 2: Background information and theory
- Chapter 3: (Laboratory) experimental setup
- Chapter 4: Details of experiment 1
- Chapter 5: Details of experiment 2
- Chapter 6: Discussion of the experimental results
- Chapter 7: Conclusion and future directions

This chapter layout is specialised for the experimental sciences, your discipline may be different.

Figures – this folder contains all figures for the thesis. These are the final images that will go into the thesis document.

6.4.2 Files

Included are also several files, most of them are plain text and you can see their contents in a text editor. After initial compilation, you will see that more auxiliary files are created by L^AT_EX or BibTeX and which you don't need to delete or worry about:

example.bib – this is an important file that contains all the bibliographic information and references that you will be citing in the thesis for use with BibTeX. You can write it manually, but

there are reference manager programs available that will create and manage it for you. Bibliographies in \LaTeX are a large subject and you may need to read about BibTeX before starting with this. Many modern reference managers will allow you to export your references in BibTeX format which greatly eases the amount of work you have to do.

MastersDoctoralThesis.cls – this is an important file. It is the class file that tells \LaTeX how to format the thesis.

main.pdf – this is your beautifully typeset thesis (in the PDF file format) created by \LaTeX . It is supplied in the PDF with the template and after you compile the template you should get an identical version.

main.tex – this is an important file. This is the file that you tell \LaTeX to compile to produce your thesis as a PDF file. It contains the framework and constructs that tell \LaTeX how to layout the thesis. It is heavily commented so you can read exactly what each line of code does and why it is there. After you put your own information into the *THESIS INFORMATION* block – you have now started your thesis!

Files that are *not* included, but are created by \LaTeX as auxiliary files include:

main.aux – this is an auxiliary file generated by \LaTeX , if it is deleted \LaTeX simply regenerates it when you run the main **.tex** file.

main.bbl – this is an auxiliary file generated by BibTeX, if it is deleted, BibTeX simply regenerates it when you run the **main.aux** file. Whereas the **.bib** file contains all the references you have, this **.bbl** file contains the references you have actually cited in the thesis and is used to build the bibliography section of the thesis.

main.blg – this is an auxiliary file generated by BibTeX, if it is deleted BibTeX simply regenerates it when you run the main **.aux** file.

main.lof – this is an auxiliary file generated by \LaTeX , if it is deleted \LaTeX simply regenerates it when you run the main **.tex** file. It tells \LaTeX how to build the *List of Figures* section.

main.log – this is an auxiliary file generated by \LaTeX , if it is deleted \LaTeX simply regenerates it when you run the main **.tex** file. It contains messages from \LaTeX , if you receive errors and warnings from \LaTeX , they will be in this **.log** file.

main.lot – this is an auxiliary file generated by \LaTeX , if it is deleted \LaTeX simply regenerates it when you run the main **.tex** file. It tells \LaTeX how to build the *List of Tables* section.

main.out – this is an auxiliary file generated by L^AT_EX, if it is deleted L^AT_EX simply regenerates it when you run the main `.tex` file.

So from this long list, only the files with the `.bib`, `.cls` and `.tex` extensions are the most important ones. The other auxiliary files can be ignored or deleted as L^AT_EX and BibTeX will regenerate them.

6.5 Filling in Your Information in the `main.tex` File

You will need to personalise the thesis template and make it your own by filling in your own information. This is done by editing the `main.tex` file in a text editor or your favourite LaTeX environment.

Open the file and scroll down to the third large block titled *THESIS INFORMATION* where you can see the entries for *University Name*, *Department Name*, etc ...

Fill out the information about yourself, your group and institution. You can also insert web links, if you do, make sure you use the full URL, including the `http://` for this. If you don't want these to be linked, simply remove the `\href{url}{name}` and only leave the name.

When you have done this, save the file and recompile `main.tex`. All the information you filled in should now be in the PDF, complete with web links. You can now begin your thesis proper!

6.6 The `main.tex` File Explained

The `main.tex` file contains the structure of the thesis. There are plenty of written comments that explain what pages, sections and formatting the L^AT_EX code is creating. Each major document element is divided into commented blocks with titles in all capitals to make it obvious what the following bit of code is doing. Initially there seems to be a lot of L^AT_EX code, but this is all formatting, and it has all been taken care of so you don't have to do it.

Begin by checking that your information on the title page is correct. For the thesis declaration, your institution may insist on something different than the text given. If this is the case, just replace what you see with what is required in the *DECLARATION PAGE* block.

Then comes a page which contains a funny quote. You can put your own, or quote your favourite scientist, author, person, and so on. Make sure to put the name of the person who you took the quote from.

Following this is the abstract page which summarises your work in a condensed way and can almost be used as a standalone document to describe what you have done. The text you write will cause the heading to move up so don't worry about running out of space.

Next come the acknowledgements. On this page, write about all the people who you wish to thank (not forgetting parents, partners and your advisor/supervisor).

The contents pages, list of figures and tables are all taken care of for you and do not need to be manually created or edited. The next set of pages are more likely to be optional and can be deleted since they are for a more technical thesis: insert a list of abbreviations you have used in the thesis, then a list of the physical constants and numbers you refer to and finally, a list of mathematical symbols used in any formulae. Making the effort to fill these tables means the reader has a one-stop place to refer to instead of searching the internet and references to try and find out what you meant by certain abbreviations or symbols.

The list of symbols is split into the Roman and Greek alphabets. Whereas the abbreviations and symbols ought to be listed in alphabetical order (and this is *not* done automatically for you) the list of physical constants should be grouped into similar themes.

The next page contains a one line dedication. Who will you dedicate your thesis to?

Finally, there is the block where the chapters are included. Uncomment the lines (delete the % character) as you write the chapters. Each chapter should be written in its own file and put into the *Chapters* folder and named **Chapter1**, **Chapter2**, etc... Similarly for the appendices, uncomment the lines as you need them. Each appendix should go into its own file and placed in the *Appendices* folder.

After the preamble, chapters and appendices finally comes the bibliography. The bibliography style (called *authoryear*) is used for the bibliography and is a fully featured style that will even include links to where the referenced paper can be found online. Do not underestimate how grateful your reader will be to find that a reference to a paper is just a click away. Of course, this relies on you putting the URL information into the BibTeX file in the first place.

6.7 Thesis Features and Conventions

To get the best out of this template, there are a few conventions that you may want to follow.

One of the most important (and most difficult) things to keep track of in such a long document as a thesis is consistency. Using certain conventions and ways of doing things (such as using a Todo list) makes the job easier. Of course, all of these are optional and you can adopt your own method.

6.7.1 Printing Format

This thesis template is designed for double sided printing (i.e. content on the front and back of pages) as most theses are printed and bound this way. Switching to one sided printing is as simple as uncommenting the *oneside* option of the `documentclass` command at the top of the `main.tex` file. You may then wish to adjust the margins to suit specifications from your institution.

The headers for the pages contain the page number on the outer side (so it is easy to flick through to the page you want) and the chapter name on the inner side.

The text is set to 11 point by default with single line spacing, again, you can tune the text size and spacing should you want or need to using the options at the very start of `main.tex`. The spacing can be changed similarly by replacing the *singlespacing* with *onehalfspacing* or *doublespacing*.

6.7.2 Using US Letter Paper

The paper size used in the template is A4, which is the standard size in Europe. If you are using this thesis template elsewhere and particularly in the United States, then you may have to change the A4 paper size to the US Letter size. This can be done in the margins settings section in `main.tex`.

Due to the differences in the paper size, the resulting margins may be different to what you like or require (as it is common for institutions to dictate certain margin sizes). If this is the case, then the margin sizes can be tweaked by modifying the values in the same block as where you set the paper size. Now your document should be set up for US Letter paper size with suitable margins.

6.7.3 References

The `biblatex` package is used to format the bibliography and inserts references such as this one (**Reference1**). The options used in the `main.tex` file mean that the in-text citations of references

are formatted with the author(s) listed with the date of the publication. Multiple references are separated by semicolons (e.g. (**Reference2**; **Reference1**)) and references with more than three authors only show the first author with *et al.* indicating there are more authors (e.g. (**Reference3**)). This is done automatically for you. To see how you use references, have a look at the `Chapter1.tex` source file. Many reference managers allow you to simply drag the reference into the document as you type.

Scientific references should come *before* the punctuation mark if there is one (such as a comma or period). The same goes for footnotes¹. You can change this but the most important thing is to keep the convention consistent throughout the thesis. Footnotes themselves should be full, descriptive sentences (beginning with a capital letter and ending with a full stop). The APA6 states: “Footnote numbers should be superscripted, [...], following any punctuation mark except a dash.” The Chicago manual of style states: “A note number should be placed at the end of a sentence or clause. The number follows any punctuation mark except the dash, which it precedes. It follows a closing parenthesis.”

The bibliography is typeset with references listed in alphabetical order by the first author’s last name. This is similar to the APA referencing style. To see how L^AT_EX typesets the bibliography, have a look at the very end of this document (or just click on the reference number links in in-text citations).

A Note on bibtex

The bibtex backend used in the template by default does not correctly handle unicode character encoding (i.e. "international" characters). You may see a warning about this in the compilation log and, if your references contain unicode characters, they may not show up correctly or at all. The solution to this is to use the biber backend instead of the outdated bibtex backend. This is done by finding this in `main.tex`: `backend=bibtex` and changing it to `backend=biber`. You will then need to delete all auxiliary BibTeX files and navigate to the template directory in your terminal (command prompt). Once there, simply type `biber main` and biber will compile your bibliography. You can then compile `main.tex` as normal and your bibliography will be updated. An alternative is to set up your LaTeX editor to compile with biber instead of bibtex, see [here](#) for how to do this for various editors.

¹Such as this footnote, here down at the bottom of the page.

TABLE 6.1: The effects of treatments X and Y on the four groups studied.

Groups	Treatment X	Treatment Y
1	0.2	0.8
2	0.17	0.7
3	0.24	0.75
4	0.68	0.3

6.7.4 Tables

Tables are an important way of displaying your results, below is an example table which was generated with this code:

```
\begin{table}
\caption{The effects of treatments X and Y on the four groups studied.}
\label{tab:treatments}
\centering
\begin{tabular}{l l l}
\toprule
\thead{Groups} & \thead{Treatment X} & \thead{Treatment Y} \\
\midrule
1 & 0.2 & 0.8 \\
2 & 0.17 & 0.7 \\
3 & 0.24 & 0.75 \\
4 & 0.68 & 0.3 \\
\bottomrule
\end{tabular}
\end{table}
```

You can reference tables with `\ref{<label>}` where the label is defined within the table environment. See `Chapter1.tex` for an example of the label and citation (e.g. Table 6.1).

6.7.5 Figures

There will hopefully be many figures in your thesis (that should be placed in the *Figures* folder).

The way to insert figures into your thesis is to use a code template like this:

```
\begin{figure}
```



```
\centering
\includegraphics{Figures/Electron}
\decoRule
\caption[An Electron]{An electron (artist's impression).}
\label{fig:Electron}
\end{figure}
```

Also look in the source file. Putting this code into the source file produces the picture of the electron that you can see in the figure below.



FIGURE 6.1: An electron (artist's impression).

Sometimes figures don't always appear where you write them in the source. The placement depends on how much space there is on the page for the figure. Sometimes there is not enough room to fit a figure directly where it should go (in relation to the text) and so \LaTeX puts it at the top of the next page. Positioning figures is the job of \LaTeX and so you should only worry about making them look good!

Figures usually should have captions just in case you need to refer to them (such as in Figure 6.1). The `\caption` command contains two parts, the first part, inside the square brackets is the title

that will appear in the *List of Figures*, and so should be short. The second part in the curly brackets should contain the longer and more descriptive caption text.

The `\decoRule` command is optional and simply puts an aesthetic horizontal line below the image. If you do this for one image, do it for all of them.

L^AT_EX is capable of using images in pdf, jpg and png format.

6.7.6 Typesetting mathematics

If your thesis is going to contain heavy mathematical content, be sure that L^AT_EX will make it look beautiful, even though it won't be able to solve the equations for you.

The “Not So Short Introduction to L^AT_EX” (available on [CTAN](#)) should tell you everything you need to know for most cases of typesetting mathematics. If you need more information, a much more thorough mathematical guide is available from the AMS called, “A Short Math Guide to L^AT_EX” and can be downloaded from: <ftp://ftp.ams.org/pub/tex/doc/amsmath/short-math-guide.pdf>

There are many different L^AT_EX symbols to remember, luckily you can find the most common symbols in [The Comprehensive L^AT_EX Symbol List](#).

You can write an equation, which is automatically given an equation number by L^AT_EX like this:

```
\begin{equation}
E = mc^2
\label{eqn:Einstein}
\end{equation}
```

This will produce Einstein's famous energy-matter equivalence equation:

$$E = mc^2 \tag{6.1}$$

All equations you write (which are not in the middle of paragraph text) are automatically given equation numbers by L^AT_EX. If you don't want a particular equation numbered, use the unnumbered form:

```
\[ a^2=4 \]
```

6.8 Sectioning and Subsectioning

You should break your thesis up into nice, bite-sized sections and subsections. \LaTeX automatically builds a table of Contents by looking at all the `\chapter{}`, `\section{}` and `\subsection{}` commands you write in the source.

The Table of Contents should only list the sections to three (3) levels. A `chapter{}` is level zero (0). A `\section{}` is level one (1) and so a `\subsection{}` is level two (2). In your thesis it is likely that you will even use a `subsubsection{}`, which is level three (3). The depth to which the Table of Contents is formatted is set within `MastersDoctoralThesis.cls`. If you need this changed, you can do it in `main.tex`.

6.9 In Closing

You have reached the end of this mini-guide. You can now rename or overwrite this pdf file and begin writing your own `Chapter1.tex` and the rest of your thesis. The easy work of setting up the structure and framework has been taken care of for you. It's now your job to fill it out!

Good luck and have lots of fun!

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Appendix A

Frequently Asked Questions

A.1 How do I change the colors of links?

The color of links can be changed to your liking using:

```
\hypersetup{urlcolor=red}, or
```

```
\hypersetup{citecolor=green}, or
```

```
\hypersetup{allcolor=blue}.
```

If you want to completely hide the links, you can use:

```
\hypersetup{allcolors=}, or even better:
```

```
\hypersetup{hidelinks}.
```

If you want to have obvious links in the PDF but not the printed text, use:

```
\hypersetup{colorlinks=false}.
```

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