Musical Transcription of Drum Patterns Using Main Audio Features as Feature Vector in KNN

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Abstract—Unlike in chordal instruments, there is limited printed musical information on how to play the drum part of popular songs. This research aims to produce a musical transcription of a drum audio recording in MIDI and PDF format. Each onsets from the input was detected and extracted of their main audio features, which were fed into a KNN classifier. Classifier data set and test set came from the audio features of downloaded drum samples that were combined to produce other physiologically available drum sound classes. The classifier performed with the highest overall accuracy of 87% when k=1 for the test set evaluation. Whereas, the actual accuracy for transcribing an input of a common drum pattern was 50%.

I. Introduction

Music has the power motivate, stir emotions, and even revive a persons memory [1], and in contemporary setting, it is usually accompanied by a percussion instrument which is mainly the drums. The drums is a rhythmic percussion instrument that is composed mainly of seven main instruments, namely bass drum (kick), snare, tom-toms, floor tom, hi-hat, crash, and ride cymbal. In a band setting, it allows listeners to follow the beat of a composition and helps establish its dynamics and overall feel. It is also known as the backbone of the band along with the electric bass.

Automatic Drum Transcription (ADT) in simple terms is the process of converting a drum performance into a record usually as a drum notation printed as a music sheet.

A. Statement of the Problem

Beginner and professional drummers alike encounter difficulty in learning songs by ear and memorization. This method has been common for folk, rock, and pop musicians [2]. An alternative and more accurate method, one that is common for schooled drummers, would to read a drum transcription or chart. Although learning and familiarizing to read charts can be an easy skill to pickup, writing charts takes more time and effort. This challenge encourages drummers to practice the former method, but people tend to forget the correct parts and play inconsistently.

Unlike charts with chords and lyrics for chordal instruments, most popular songs do not have available drum transcriptions, e.g., Figure 1. It would also be difficult and time-consuming for beginners to write charts themselves. But, there is a

Presented to the Faculty of the Institute of Computer Science, University of the Philippines Los Baños in partial fulfillment of the requirements for the Degree of Bachelor of Science in Computer Science

convenient way to electronically transcribe drums music. It is by using a musical instrument digital interface (MIDI) controller, connected to a Digital Audio Workstation (DAW) that is installed on a computer. However convenient, such devices may not always be practical for common drummers.

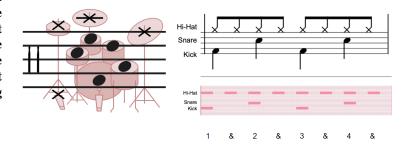


Fig. 1. Sample notation guide, and drum transcription of the song Billy Jean and its midi visualization [18], [19].

B. Significance of the Study

Since there are very few songs that have written drum patterns for another player to learn, this study suggests a solution to automate the audio to musical transcription.

C. Objectives

The general objective of this study was to develop an application which can convert a digital drum recording into a musical drum transcription.

Specific:

- To produce physiologically realistic drum sound combinations using downloaded drum instrument audio samples
- 2) To extract main audio features from the data set
- 3) To classify the onsets in a given drum recording
- To produce a PDF and MIDI transcription out of the classified onsets
- 5) To evaluate the accuracy of the produced transcription

II. SCOPE AND LIMITATION

The audio input tempo and beat can be tracked using the Librosa package for common time signatures, but it is not always successful when given polyrhythmic beats. The detected tempo is sometimes double of the original, thus, the correct tempo was made as an optional argument to the program. Beat onsets or hits are also detected, but misses a few when not loud enough.

Another concern for transcription was finding the downbeat, or the time where to count "1" in a song, as there are no solutions yet to this problem. Thus, for this work, the first onset in the input recording should be the downbeat, and the time signature should be 4/4, although is already common in playing drums.

The extracted audio feature dimension was reduced. Some where truncated while some were averaged to form 77-dimensional feature array. The truncation was done to reduce the computational complexity of KNN, and to produce a standard feature length with respect to the start the note or onset, regardless of the note length.

A manual evaluation to compare the classes input vs output classes was used as there is not yet a solution for music similarity problem.

III. RELATED LITERATURE

Studies on music transcription have mostly been interested on melodic instruments, but there were still some that focused on ADT, mostly using audio features. [3] The use of audio features was common in automatic music transcription (AMT) [12]. This is also used in ADT, but in a different implementation.

FitzGerald, identifies ADT as a sound source separation problem. His approach of using simple heuristics was able to identify individual drum parts without any form of rhythmic modelling, where he used the general source separation and redundancy reduction techniques, such as Independent Component Analysis and Independent Subspace Analysis. There were only a few issues of misidentifying a low tom as a kick drum, and a loud hi-hat being detected as a drum hit [5].

Southall, Stables and Hockman used Convolutional Neural Network (CNN) in their implementation of ADT. They used spectral features, producing a segmented spectrogram to be the training data for CNN. Their model was trained to identify only the kick drum, snare drum, and hi-hat sounds. It was evaluated using a drum solo, which is a drum mixture (with drum set parts other than kick, snare, and hi-hat), and with a multi-instrument mixture for the purpose of testing their systems ability to adapt from unseen timbres [6].

The study of Miron, Davies, and Gouyon [13] also used feature extraction as feature vectors and the K-Nearest Neighbor algorithm (KNN) for ADT, and they were able to produce an almost 90% accuracy in recognizing the sounded instruments, but the algorithm was intended for live input and only short samples.

In the study of Gillet and Richard, they performed ADT from an audio-video file by extracting its audio and visual features. They compared the recognition rate when using only the audio features, only the video features, and joint feature vectors. They found out the using the joint feature vectors best among the three [7].

A methodology simmilar to an undergraduate study [14] was adopted for this study because of their same case of using KNN to classify audio signals using low-level features, but the study focused on heartbeat classification. This study suggested to increase the sample size to improve classification performance.

IV. METHODOLOGY

This study focused on classifying drum onsets with respect to time, then produce the transcription. This study was composed of four major processes: data preparation, feature extraction, classification, and transcription. These were implemented using Python 3. The process flow can be found in Figure 2.

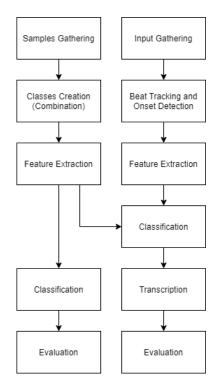


Fig. 2. Methodology

A. Data Gathering

There are 7 main kinds instruments present in a drum kit, namely bass, snare, tom-tom, floor tom, hi-hat, crash, and ride, where three more sounds may be produced, namely open hihat, stick, and ride bell, for a total 10 different sounds. Samples of each were downloaded from several websites but mostly from [22], which provided the compilation of initially labeled samples for the single hit classes. All samples were in WAV format of no more than 3 seconds. The single hit class representatives were combined to produce a total of 142 classes, which constituted all the data set. About 500 representatives were chosen from each class. The single class samples were duplicated up to a total of about 500. This was done to balance the weight of samples for KNN. Meanwhile, other derived classes' samples grew exponentially. The script for the combination necessary for producing classes was included in the program.

The single hit classes were combined in reference to the drummer's physiology as seen in Table 1. The complete enumeration of classes is found in Appendix I.

Table 1: Breakdown of the derived classes

Instrument	Derivation (count)	Subtotal
count per		
beat		
1	kit pieces (7) + stick, hhopen	10
	and bell (3)	
2	two sticks (36) + one stick	45
	and bass (9)	
3	two sticks and bass (48) +	68
	two sticks and foot hihat (20)	
4	Two sticks, foot hihat and	19
	bass (19)	
total		142
classes		

B. Feature Extraction

Feature extraction is the process of creating a new set of k-dimenstional features from the original data [4]. It is done to reduce the computational complexity of signal processing algorithms [9]. The feature set used for classification were extracted using LibROSA [20], a Python package for music and audio analysis.

The feature set was composed of low-level audio features which were readily computed abstractions that LibROSA provides for processing raw audio inputs. The following features were used:

- 1) Spectral bandwidth describes the density of the signal. It can be computed by calling the function librosa.feature.spectral_bandwidth(y), where y is the audio time series array.
- 2) Spectral contrast describes the difference of the high and low points of the audio signals spectrum, computed as librosa.feature.spectral_contrast(y).
- 3) Zero crossing rate describes how often the audio signal crossed the 0 voltage axis, computed as librosa.feature.zero_crossing_rate(y).
- 4) *Mel-Frequency Cepstral Coefficient* which represents the shape of the spectrum, as how a human auditory perception would react with very few coefficients, computed as librosa.feature.mfcc(y)
- 5) *Spectral centroid* which describes the central mass of the input signal, computed by calling the function librosa.feature.spectral_bandwidth(y).
- 6) Spectral flatness is the quantification of a sound from being noise-like versus being tone-like, computed as librosa.feature.spectral_flatness(y)
- 7) Spectral roll-off describes the distribution of the power along the frequency spectrum, and is computed as librosa.feature.spectral_rolloff(y)

Librosa returned the features as Numpy array. The MFCC and spectral contrast were averaged as an array of length 20 and 7 respectively, while the remaining features were truncated to be 10 elements long, forming a total of 77-dimensional feature array.

C. Classification

The algorithm used the Scikit-learn package for KNN [21] performed classification over the data set. 10% of the data was

partitioned as the validation set. The evaluation for classifying individual samples was done using a built in abstraction of the package for evaluating KNN accuracy.

3

KNN decreases in performance as it increases the feature dimension, so the data set size was increased to compensate [16]. A large sample size of about 500 for each class was used.

The algorithm was utilized to perform the classification for longer audio streams of drum beat patterns. The audio input was segmented by the the number of its onsets using the Librosa onset detection abstraction. Onset positions was recorded with reference to beat locations using the Librosa beat tracker. Each onsets was fed into the classifier. Onset classifications with respect to tracked beats were transcribed using a script included in the program. The output was produced as a MIDI format file using the Python MIDIUtil library, and a PDF format using Lilypond [23].

D. Evaluation

The algorithm used for classifying the sounds was evaluated for its accuracy in correctly identifying the sounding classes. The Scikit-learn has a built-in library for evaluating KNN accuracy.

Also, the full transcription algorithm was run and evaluated using two types of input:

- A briefly composed midi drums played through LMMS (digital audio workstation) and imported as a wav audio file
- 2) A recorded solo drums performance

A manual comparison was done, through counting the different classes, to compare the actual classes from the input audio, versus those from the output transcription. By this method, confusion matrix was created from examining the output transcription. The confusion matrix was used to visualize and compute the precision, recall, and accuracy of the algorithm. Precision describes the percentage of relevant results being correctly classified, while recall is the percentage of relevant results. Accuracy, on the other hand, describes how correct are the results. Figure 3 shows the formulas for their computation. [17]

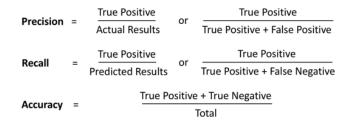


Fig. 3. Formula for computing precision, recall, and accuracy [17].

V. RESULTS AND DISCUSSIONS

The algorithm was successful in combining basic drums audio samples to produce more physiologically realistic sample classes. Out of the samples, main audio features were extracted for each. Since the combination took about 3 hours and 10 GB of memory to complete in a single process script, the

produced samples were excluded from the source. However, the extracted features were included in the source code as comma-separated values (CSV). The CSV was for loaded every transcription run, instead of recombining or reextracting audio and features.

A. K-Nearest Neighbor

The KNN classifier which used the audio features was evaluated with the highest accuracy of 98% when identifying just three classes instead of the full 142, and when k=1. The classifier was also evaluated in varying constraints—different number of cases considered, and different values of k—and getting the average of 5 test runs each. Table 2 shows their accuracy.

Table 2: KNN accuracy in varying k value and number of

Classes	k=1	k=3	k=5	k=7	k=9
Simple	98.29%	95.44%	93.15%	91.41%	91.68%
Common	97.03%	93.59%	90.66%	88.26%	86.54%
All	87.41%	75.40%	68.35%	63.91%	60.82%

The considered classes were adjustable through whitelisting or blacklisting a sound, or removing specific combinations. The *simple* class on the table pertains to the sound classes bass, snare and hihat which are the most commonly heard in pop songs. The *common* class pertains to most parts of the drum kit, except that the least common sound combinations were eliminated. The *all* class pertains to the total physiologically possible drum sound combinations.

B. Transcription

Two live and one digital drum tracks were used as input to test the performance of the transcription algorithm. The table 3 and 4 shows results of the former which use bass, snare and hihat patterns; while table 5 shows the result of the later which uses rhythm patterns of bass, snare, hihat, open-hihat, tom, floor, crash, and ride. The PDF output for tables 3 and 4 are shown on appendix 2.

Table 3: Confusion matrix of bass, snare and hihat transcriptions (truth data vs classifier results¹)

transcriptions (train data vs classifier results)					
Class	bass	snare	hihat		
bass	8	0	5		
snare	0	4	0		
hihat	2	4	4		
Precision	80%	50%	44.44%		
Recall	61.54%	100%	40%		

Ovarall accuracy 59.26%

Table 4: Confusion matrix of bass, snare and hihat transcriptions (truth data vs classifier results¹)

Class	bass	snare	hihat	
bass	16	1	5	
snare	4	18	0	
hihat	5	23	51	
Precision	64%	94.77%	91.07%	
Recall	72.73%	81.82%	91.07%	

¹in left column and top row respectively

Ovarall accuracy 85%

Table 5: Confusion matrix for commonly used classes (truth data vs classifier results)

Class	bs	sn	hh	hho	tm	fĺ	cr	rd
bass	22	7	5	1	2	3	2	1
snare	5	15	1	0	0	1	1	0
hihat	0	7	18	1	0	0	0	0
hhopn	0	0	0	7	0	0	0	1
tom	2	0	0	0	0	1	0	1
floor	2	0	0	0	2	0	0	0
crash	0	0	0	1	0	0	3	0
ride	0	1	16	0	0	0	2	2
Recall	51%	65%	69%	88%	0%	0%	75%	10%
Precisn	71%	50%	45%	70%	0%	0%	38%	40%

Ovarall accuracy 50.38%

Compared with the evaluation of the initial KNN test set, the actual performance when using the input recordings was less accurate. Still, the application results were good since most of the precision and recall levels were above 50%.

Two main factors were considered to affect the classifier performance: the sampling environment and the drum audio features themselves. The KNN dataset samples were recorded from a controlled environment, usually to minimize ambient noise. Whereas, the recorded inputs for transcription had more reverberation, thus affecting the succeeding onsets on another class. Also, the input being a continuous stream, the features of the currently evaluated class was affected by the previous class, whose features leaked to the current. This was most evident for recordings with fast tempo. In addition, the segmentation of the input based on individual onsets were not as aligned as the controlled samples. The second factor was that drum set sounds have similar features, so depending on the audio dynamics and how the audio levels were mixed, when some audio classes were loud enough, other audio features got overwhelmed by other classes, and thus missed.

VI. CONCLUSION

The program successfully combined the drum samples to produce other physiologically valid classes.

Audio features were successfully extracted from every class samples.

Using main audio features in KNN is effective in identifying individual onset types from another, attaining the highest overall accuracy of 98% for k=1 and when considering only between 3 classes, even with the feature vector dimension of 77, in this case.

The algorithm was able to output same transcriptions but in both PDF and MIDI format.

The transcription performance was evaluated with an accuracy of 50% for k=1 when considering the commonly used classes, and the highest of 85% when considering just the three classes.

VII. RECOMMENDATION

Just like speech recognition, ADT may improve with the help of other machine learning techniques like Hidden Markov Model or Artificial Neural Networks.

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APPENDIX I ENUMERATION OF CLASSES

The following figures would enumerate classes that are appropriate to the physiology of a drummer.

Instrument pieces (7) and available sounds (3)

- 1) bass
- 2) snare
- 3) tom
- 4) floor
- 5) hihat
- 6) crash
- 7) ride
- 8) stick
- 9) hihat-open
- 10) bell

One stick and bass (9)

- 1) bass snare
- 2) bass stick
- 3) bass tom
- 4) bass floor

- 5) bass hihat
- 6) bass hihat-open
- 7) bass crash
- 8) bass ride
- 9) bass bell

Two sticks (36)

- 1) snare stick
- 2) snare tom
- 3) snare floor
- 4) snare hihat
- 5) snare hihat-open
- 6) snare crash
- 7) snare ride
- 8) snare bell
- 9) stick tom
- 10) stick floor
- 11) stick hihat
- 12) stick hihat-open
- 13) stick crash
- 14) stick ride
- 15) stick bell
- 16) tom floor
- 17) tom hihat
- 18) tom hihat-open
- 19) tom crash
- 20) tom ride
- 21) tom bell
- 22) floor hihat
- 23) floor hihat-open
- 24) floor crash
- 25) floor ride
- 26) floor bell
- 27) hihat crash
- 28) hihat ride
- 29) hihat bell
- 30) hihat-open
- 31) hihat-open crash
- 32) hihat-open ride
- 33) hihat-open bell
- 34) crash ride
- 35) crash bell
- 36) ride bell

Two sticks and bass (48)

- 1) bass snare tom
- 2) bass snare floor
- 3) bass snare hihat
- 4) bass snare hihat-open
- 5) bass snare crash
- 6) bass snare ride
- 7) bass snare bell
- 8) bass stick hihat
- 9) bass stick hihat-open
- 10) bass stick tom
- 11) bass stick floor
- 12) bass stick hihat13) bass stick hihat-open
- 14) bass stick crash

- 15) bass stick ride
- 16) bass stick bell
- 17) bass tom hihat
- 18) bass tom hihat-open
- 19) bass tom floor
- 20) bass tom hihat
- 21) bass tom hihat-open
- 22) bass tom crash
- 23) bass tom ride
- 24) bass tom bell
- 25) bass tom bell hihat
- 26) bass floor hihat
- 27) bass floor hihat-open
- 28) bass floor hihat
- 29) bass floor hihat-open
- 30) bass floor crash
- 31) bass floor ride
- 32) bass floor bell
- 33) bass hihat hihat-open
- 34) bass hihat crash
- 35) bass hihat ride
- 36) bass hihat bell
- 37) bass hihat-open crash
- 38) bass hihat-open ride
- 39) bass hihat-open bell
- 40) bass crash hihat
- 41) bass crash hihat-open
- 42) bass crash ride
- 43) bass crash bell
- 44) bass ride hihat
- 45) bass ride hihat-open
- 46) bass ride bell
- 47) bass bell hihat
- 48) bass bell hihat-open

Two sticks and foot hihat (20)

- 1) snare stick hihat
- 2) snare tom hihat
- 3) snare floor hihat
- 4) snare crash hihat
- 5) snare ride hihat
- 6) snare bell hihat
- 7) stick tom hihat
- 8) stick floor hihat
- 9) stick crash hihat
- 10) stick ride hihat
- 11) stick bell hihat
- 12) tom floor hihat
- 13) tom crash hihat
- 14) tom ride hihat
- 15) tom bell hihat
- 16) floor crash hihat
- 17) floor ride hihat
- 18) floor bell hihat
- 19) crash ride hihat
- 20) crash bell hihat
- 21) ride bell hihat
- Two sticks, foot hihat and bass (19)

- 1) bass snare tom hihat
- 2) bass snare floor hihat
- 3) bass snare crash hihat
- 4) bass snare ride hihat
- 5) bass snare bell hihat
- 6) bass stick tom hihat
- 7) bass stick floor hihat
- 8) bass stick crash hihat
- 9) bass stick ride hihat
- 10) bass stick bell hihat
- 11) bass tom floor hihat
- 12) bass tom crash hihat
- 13) bass tom ride hihat
- 14) bass floor crash hihat
- 15) bass floor ride hihat
- 16) bass floor bell hihat
- 17) bass crash ride hihat
- 18) bass crash bell hihat
- 19) bass ride bell hihat

APPENDIX II TRANSCRIPTION

Shows the comparison of the correct transcription of a input vs the actual output transcription.

bass_snare_hihat_loop



Fig. 4. Correct transcription of table 3 bass_snare_hihat_loop.wav

bass_snare_hihat_loop



Fig. 5. Actual transcription of table 3 bass_snare_hihat_loop.wav

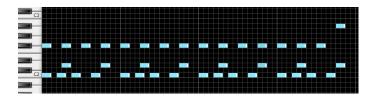


Fig. 6. Correct midi visualization of figure 4 through LMMS

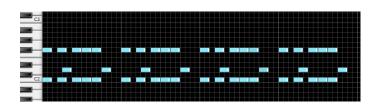


Fig. 7. Actual midi visualization of figure 5 through LMMS

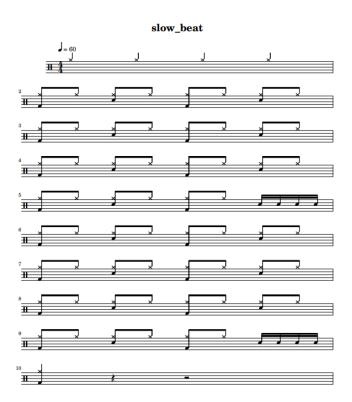
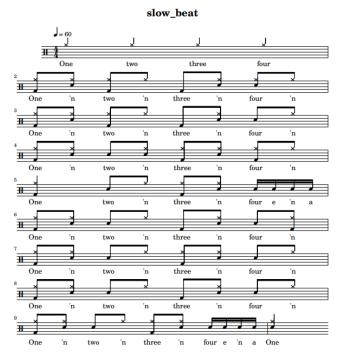


Fig. 8. Correct transcription of table 3 slow_beat.mp3



 $Fig.\ 9. \quad Actual\ transcription\ of\ table\ 3\ slow_beat.mp3$