



INCREASING DRUM TRANSCRIPTION VOCABULARY USING DATA SYNTHESIS

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Problem

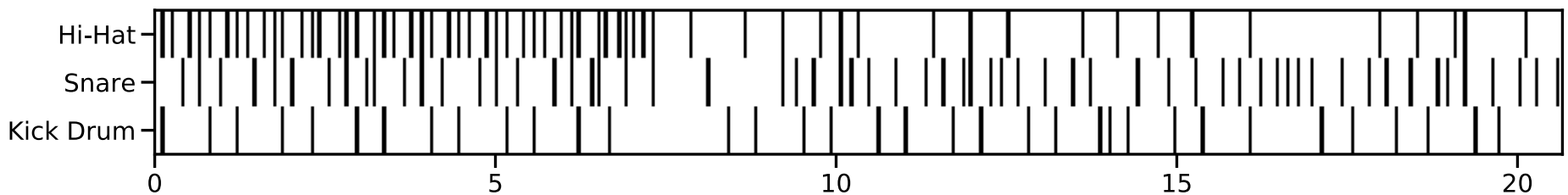
Most Automatic Drum Transcription (ADT) algorithms are limited to simply onset times of 3 classes:

1. Bass Drum (BD)
2. Snare Drum (SD)
3. Hi-Hat (HH)

Example:

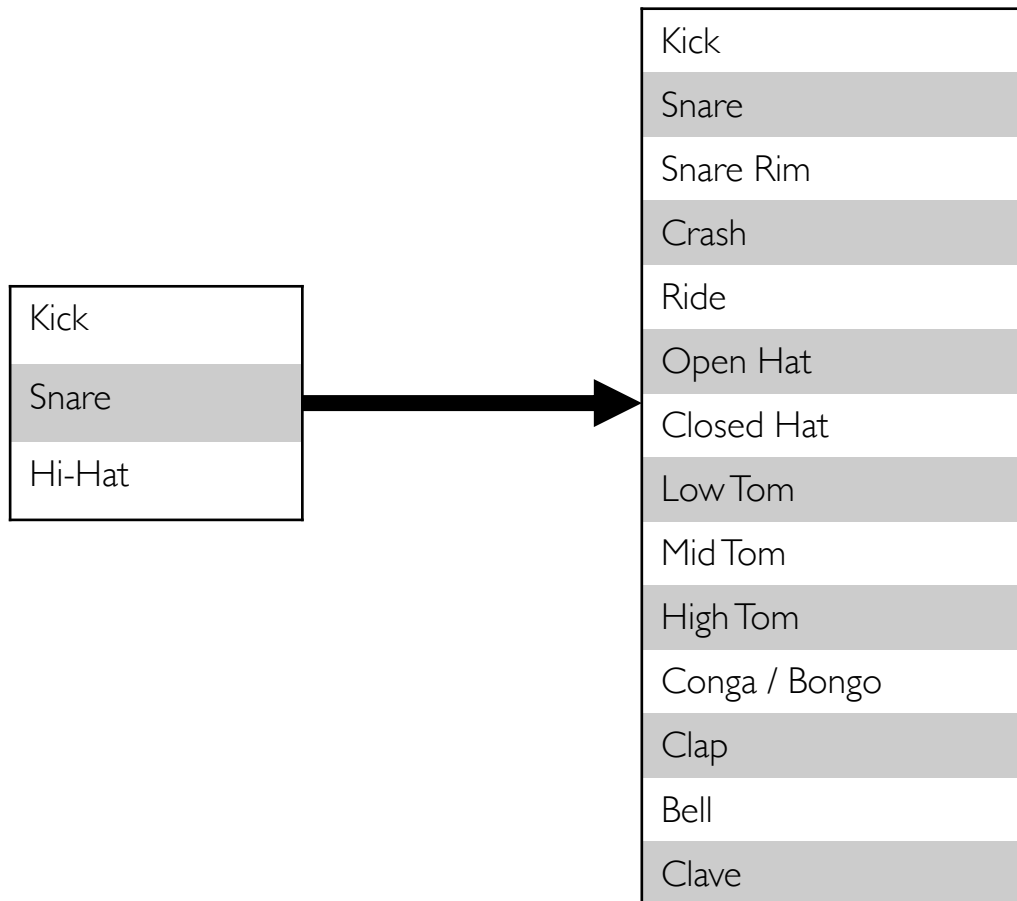
Original 

3-class re-synthesis: 

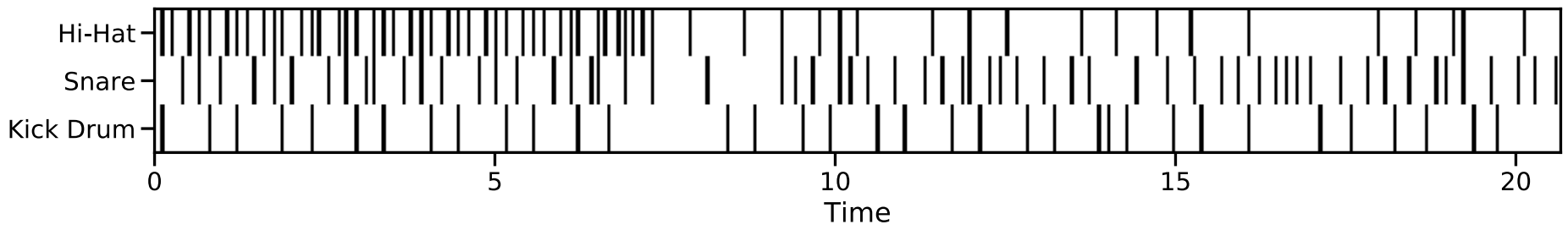


Goal

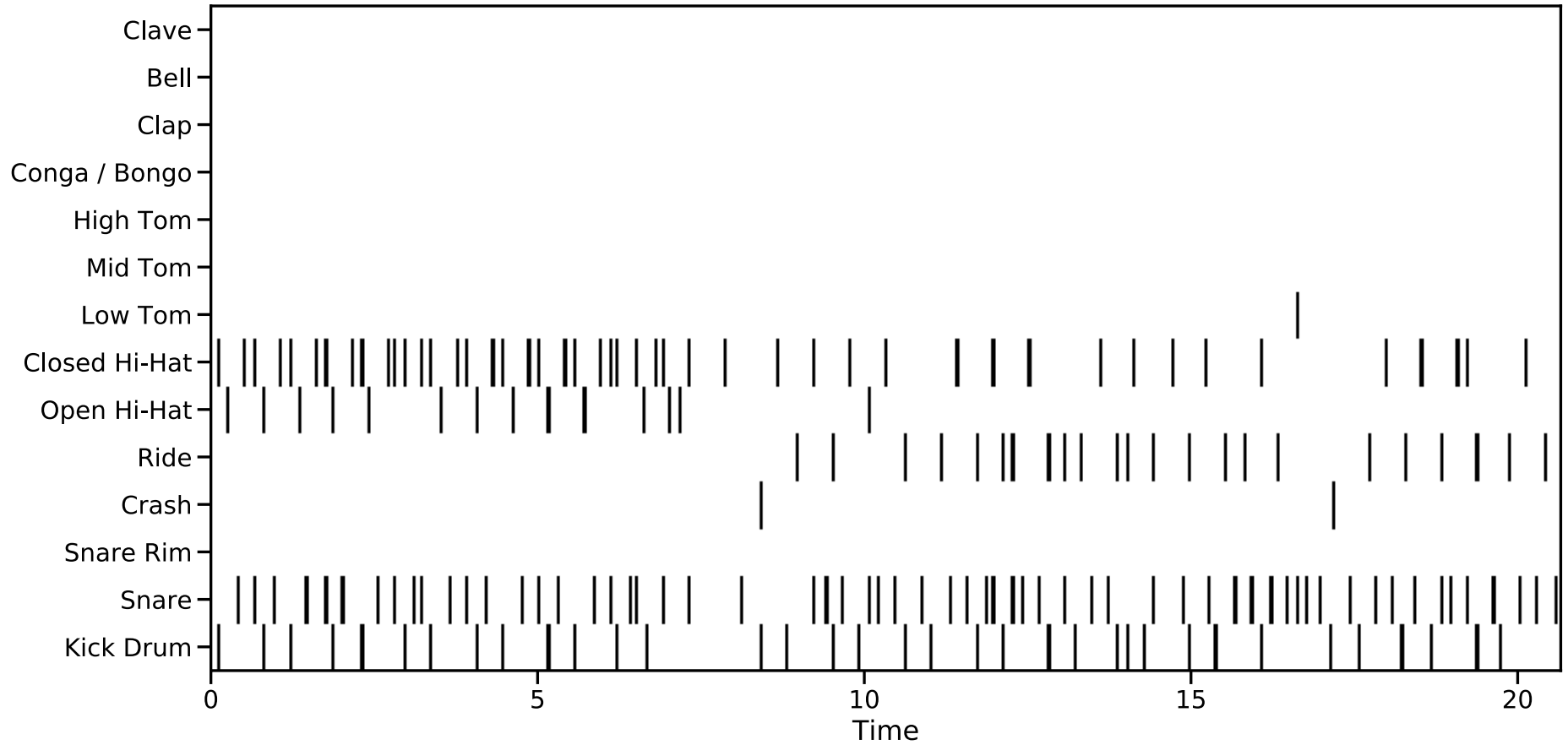
- Increase vocabulary to onset times of 14 drum classes:



Goal



Goal



ADT Datasets

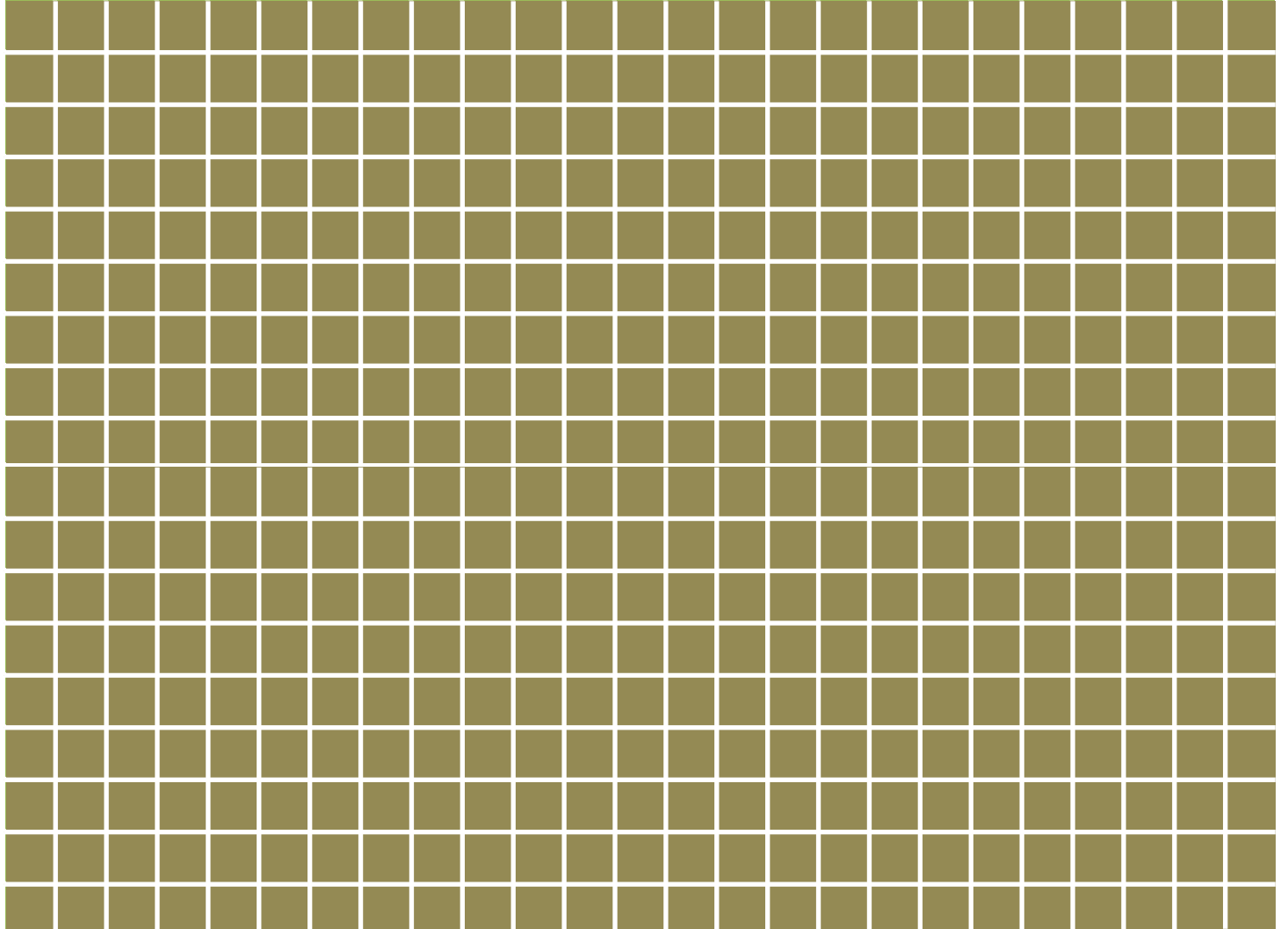


~33,000 onsets / 2 hours of 3-class data
(RBMA, IDMT/SMT)



~33,000 onsets / ~1.5 hours of > 3-class data
(ENST/MDB)

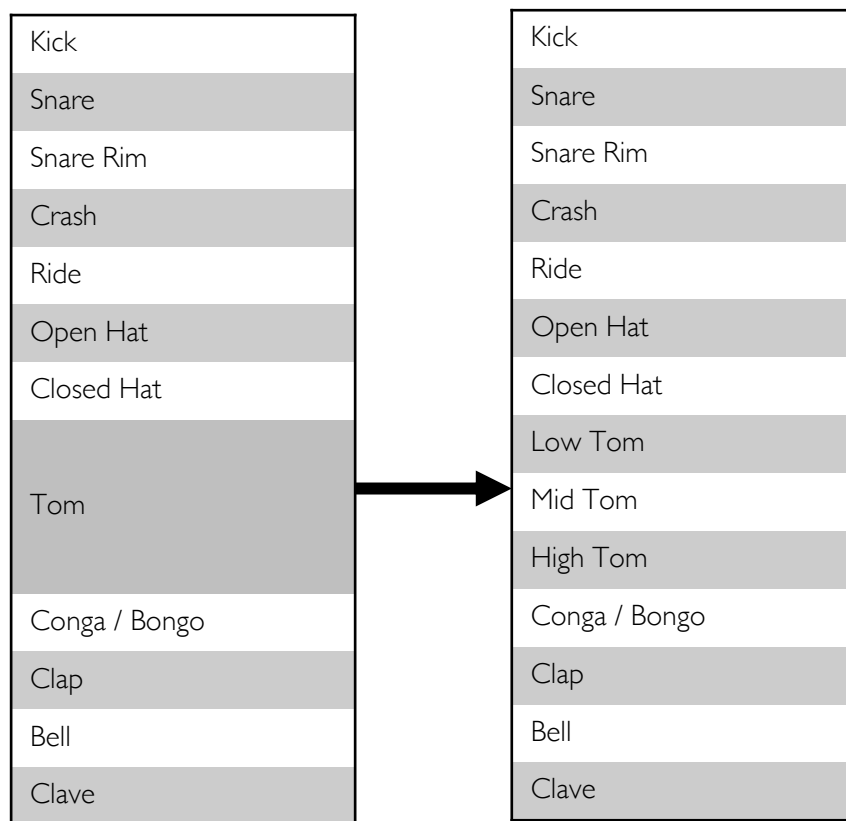
ADT Datasets



Building the S***** Drum Dataset (SDDS)

The Sounds

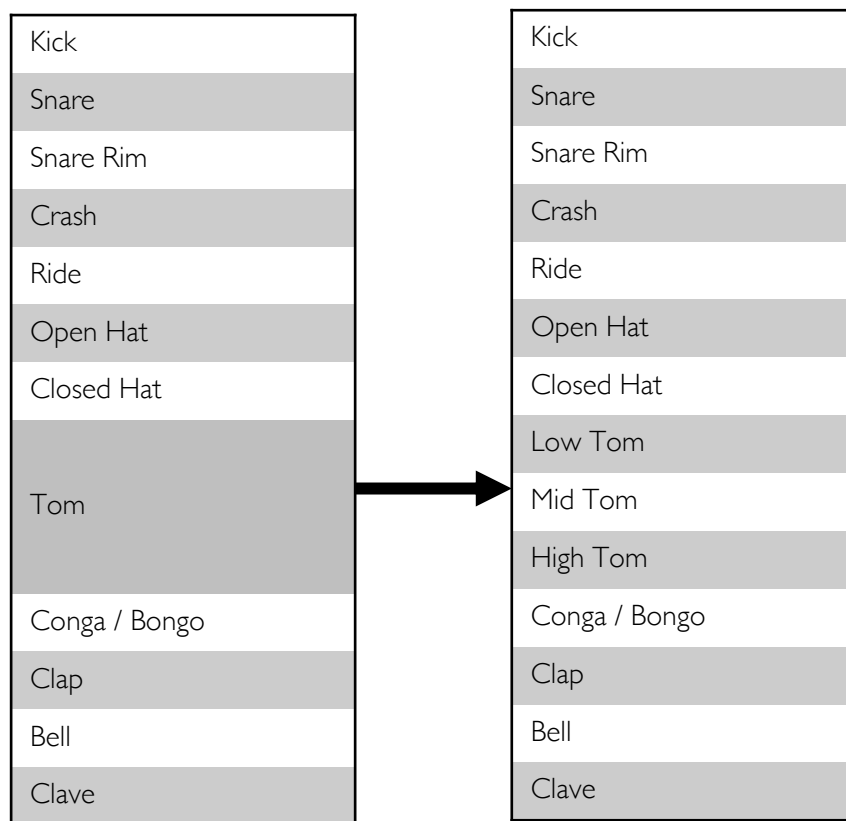
- Labeled samples from 8 drum sample libraries
- Split toms into low, mid, high based on pitch, spectral centroid
- Split dataset into 4000 train / 2000 test samples



Building the Synthetic Drum Dataset (SDDS)

The Sounds

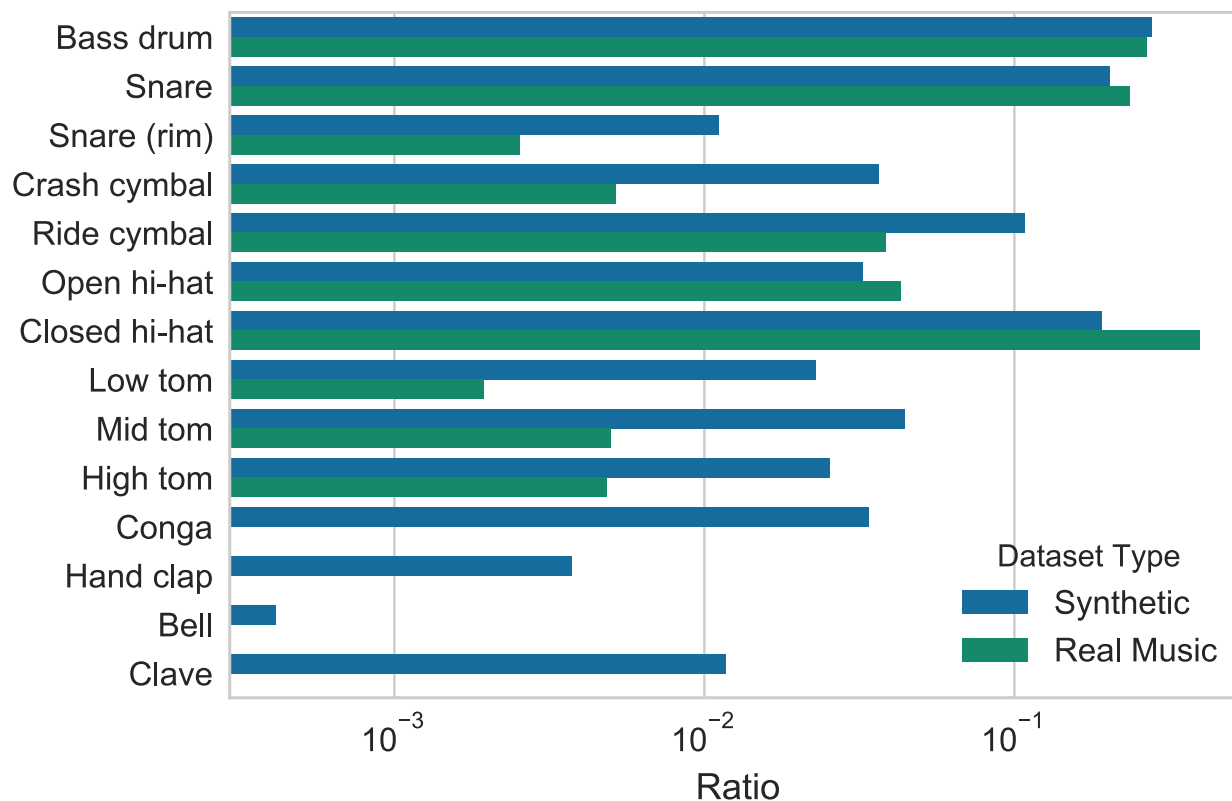
- Labeled samples from 8 drum sample libraries
- Split toms into low, mid, high based on pitch, spectral centroid
- Split dataset into 4000 train / 2000 test samples



Building the Synthetic Drum Dataset (SDDS)

The Sounds

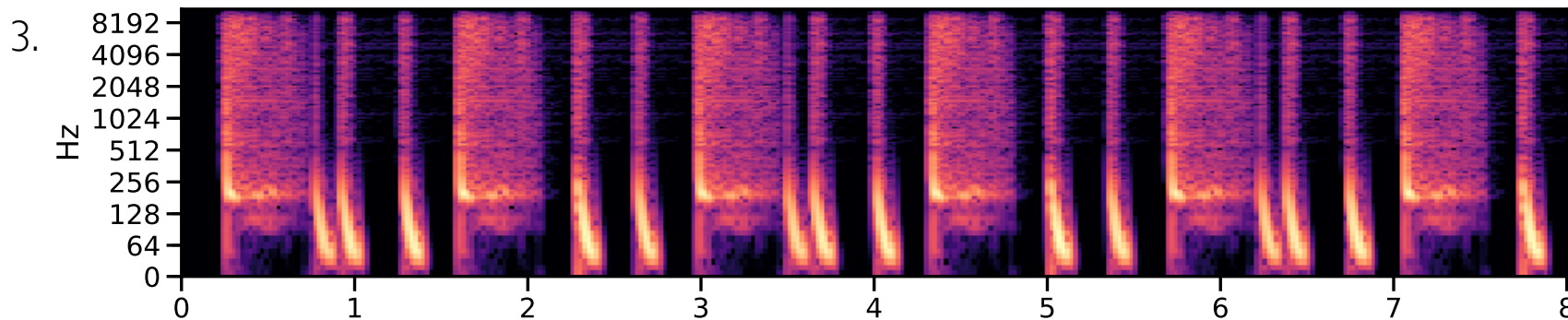
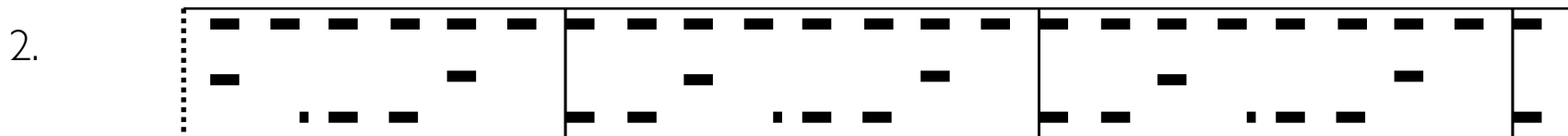
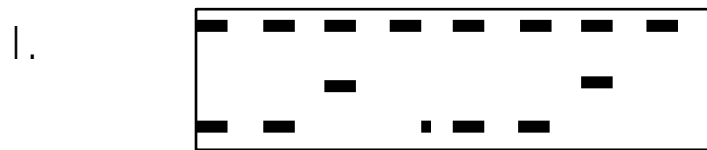
- Labeled samples from 8 drum sample libraries
- Split toms into low, mid, high based on pitch, spectral centroid
- Split dataset into 4000 train / 2000 test samples



Building the Synthetic Drum Dataset (SDDS)

The Rhythms

- 60k measures of percussion MIDI files (50k train / 5k test / 5k validate)

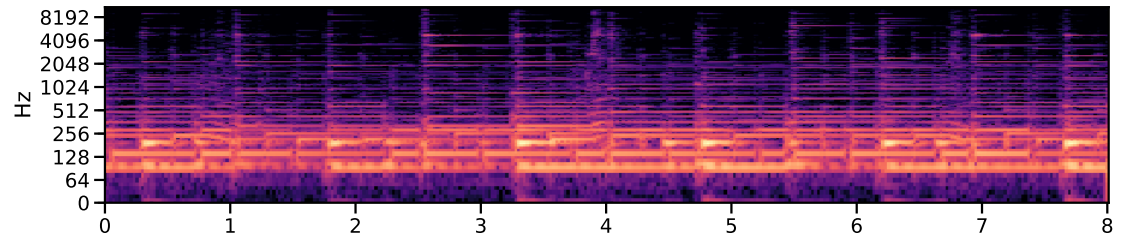


Building the Synthetic Drum Dataset (SDDS)

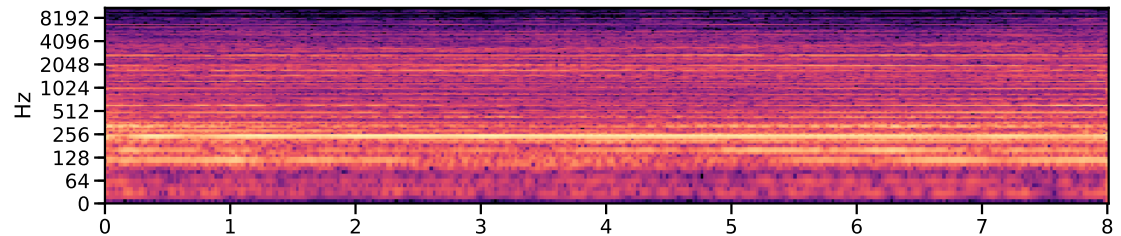
Augmentation and “Accompaniment”

- Augment to 210k (200k train / 5k test / 5k validate) w/ small pitch shifts, added pink noise, and “*harmonic noise accompaniment*”:

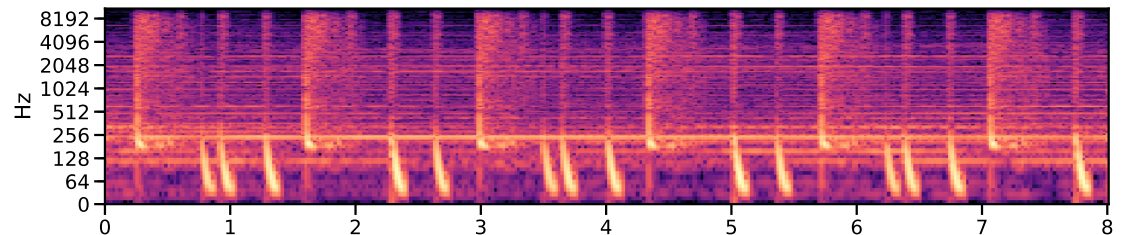
- Solo harmonic instrument recording



- “Smear” in time (fwd / bkwd reverb)



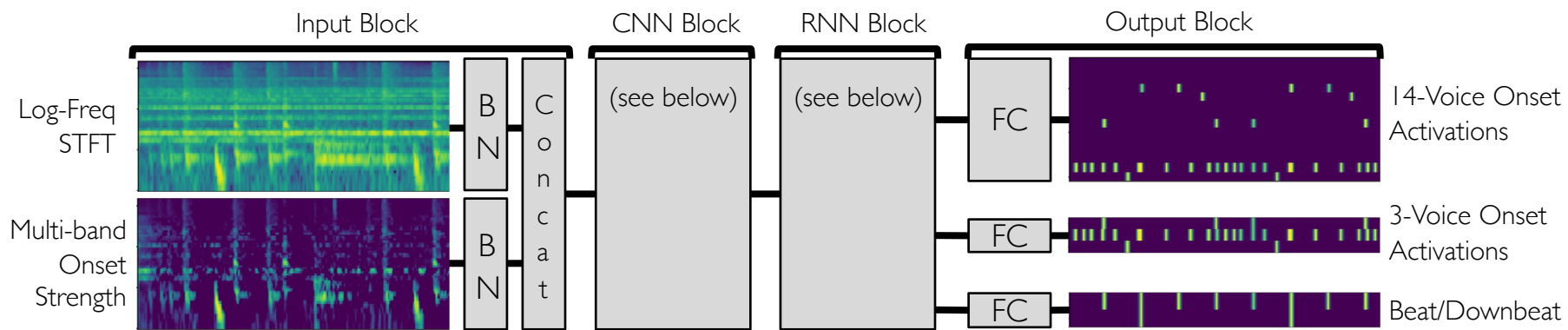
- Mix with drum track



Combined Datasets

	RBMA	SMT	ENST	MDB	SDDS
Hours	1.67	0.51	1.28	0.23	467
Accomp.	X	X		X	sort of
14-voice Onsets			X	X	X
3-voice Onsets	X	X	X	X	X
Beats	X				X

Model



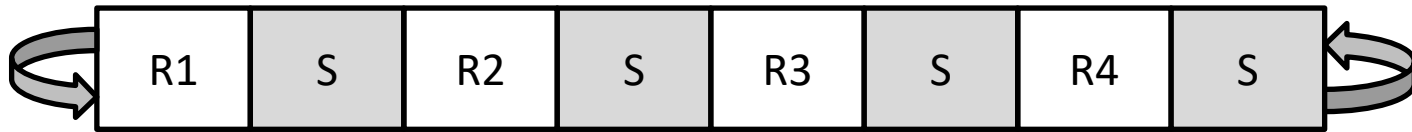
Input Features
22050 sampling rate
Log-magnitude, log-frequency STFT: 1024 samples 64 bands (8 octaves, 8 bands per octave) 40 Hz – 10kHz 0.01 sec hop
Multi-band onset strength: Computed from log-f STFT Half-rectified difference between current frame and mean of past 200 ms

CNN Block
32 (3x3) Conv
32 (3x3) Conv
Batch Norm
ReLU
30% Dropout
64 (3x3) Conv
64 (3x3) Conv
BatchNorm
ReLu
30% Dropout
64 (1x64) Conv
BatchNorm
ReLU

RNN Block
(-6:+6) Context Windowing
64-unit BLSTM
64-unit BLSTM
64-unit BLSTM

Training with Heterogeneous Outputs

- Mask outputs not in use for each example
- Use round-robin sampling with Pescador¹ to ensure all outputs used in each mini-batch of 8:



- Minimize weighted combination of binary cross-entropy losses based using weights computed by activation entropy
- 3-fold CV splits for the small real music small datasets
25% validation / 75% testing in each split

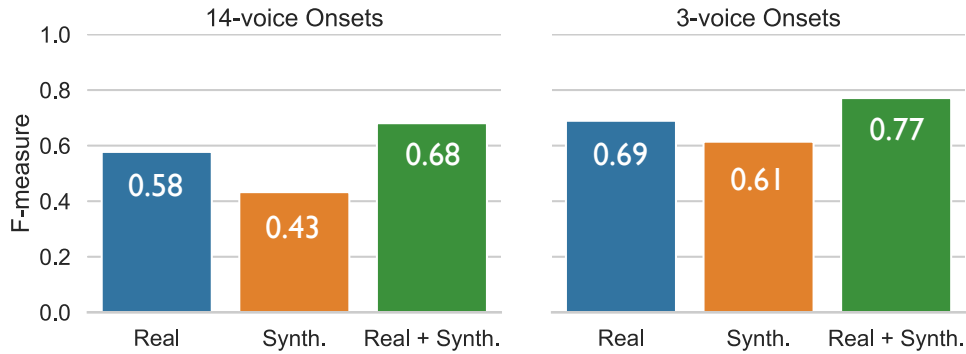
Experiments

Variables:

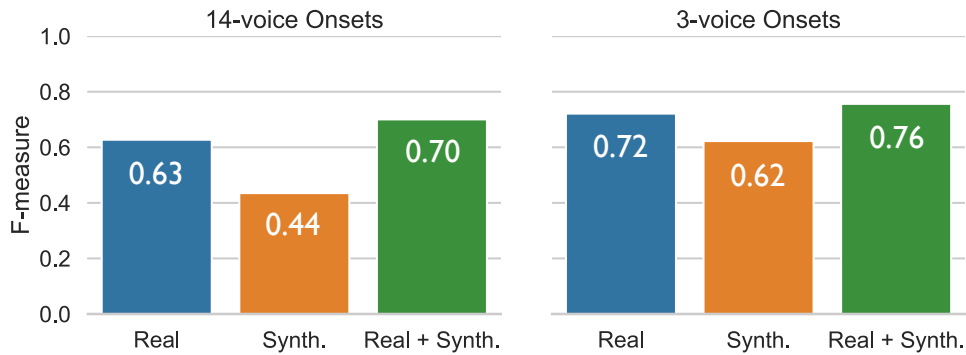
- Training data
 - Real music (RBMA, SMT, ENST, MDB)
 - Synthetic (SDDS)
 - Recorded + Synthetic
- Model capacity
 - “Small” (as described)
 - Large (more conv filters, more BLSTM units)
- Outputs
 - Multi-task
 - Single-task (w/ limited data)
- Class-weighting
 - No weighting
 - Weighted by activation entropy

Results

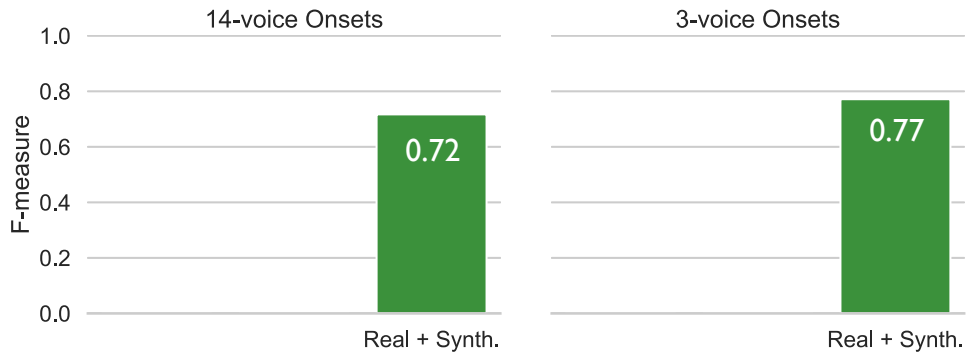
Small /
Multi-task



Large /
Multi-task

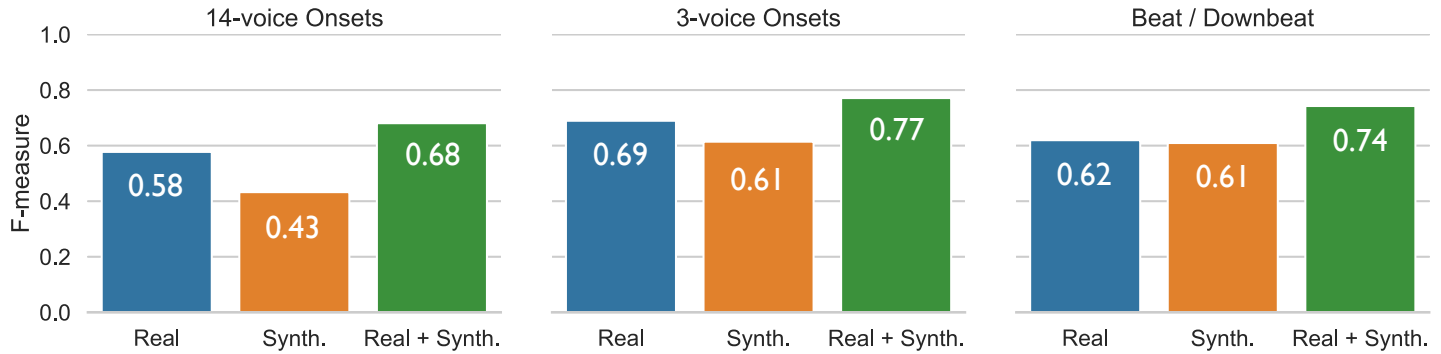


Small /
Single-task

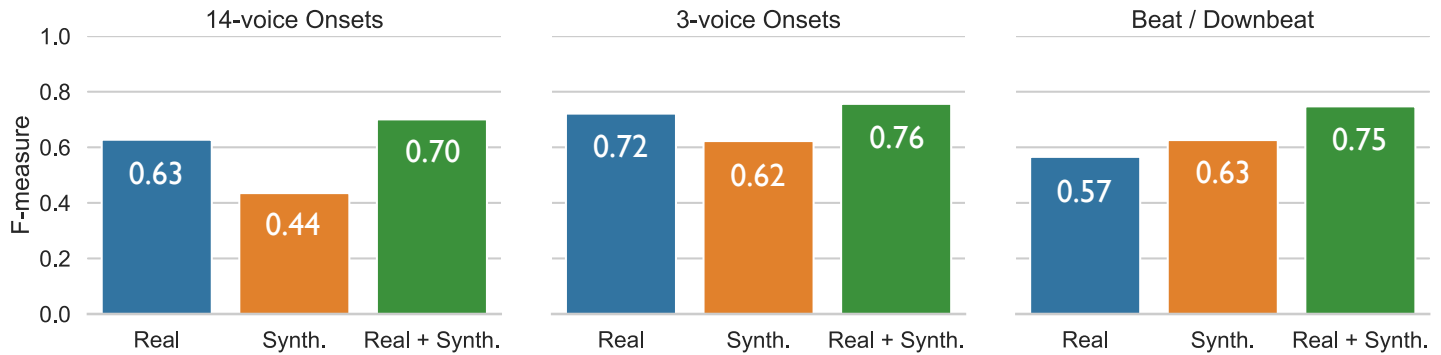


Results

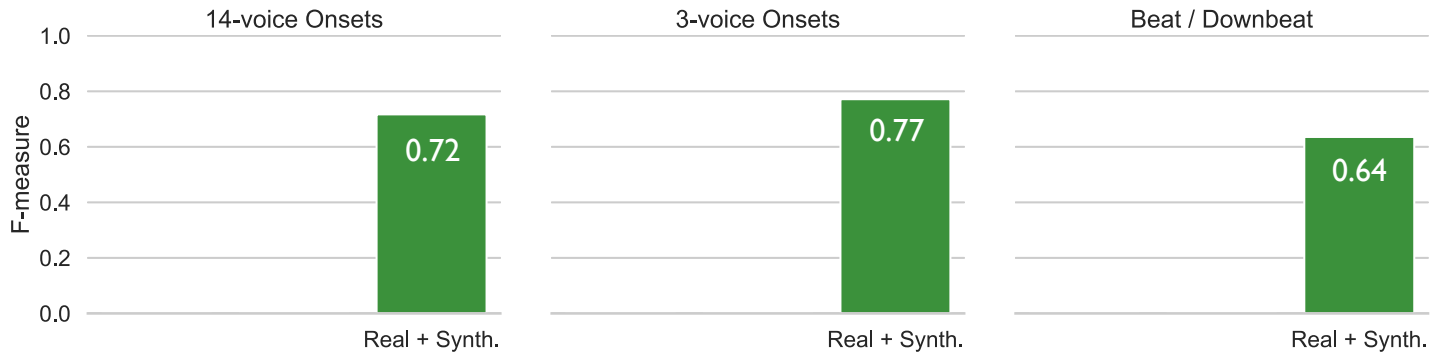
Small /
Multi-task



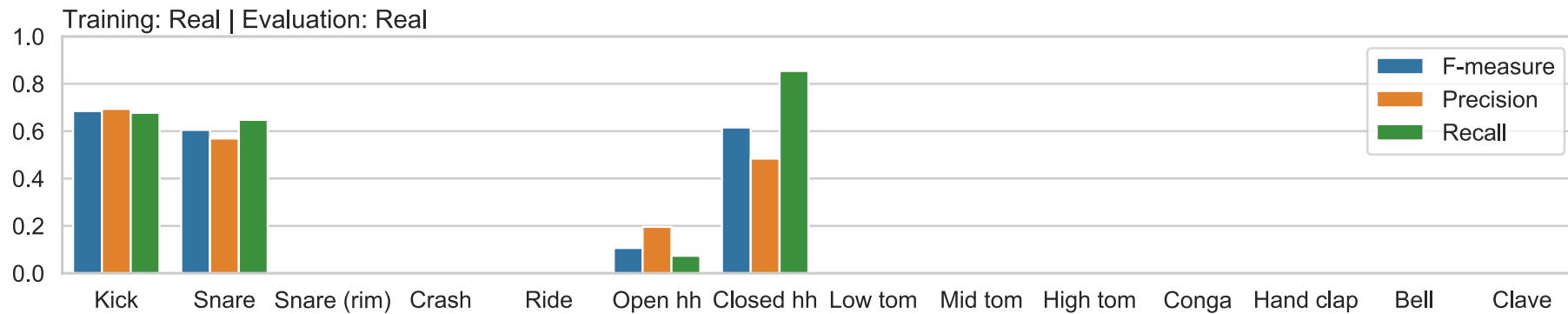
Large /
Multi-task



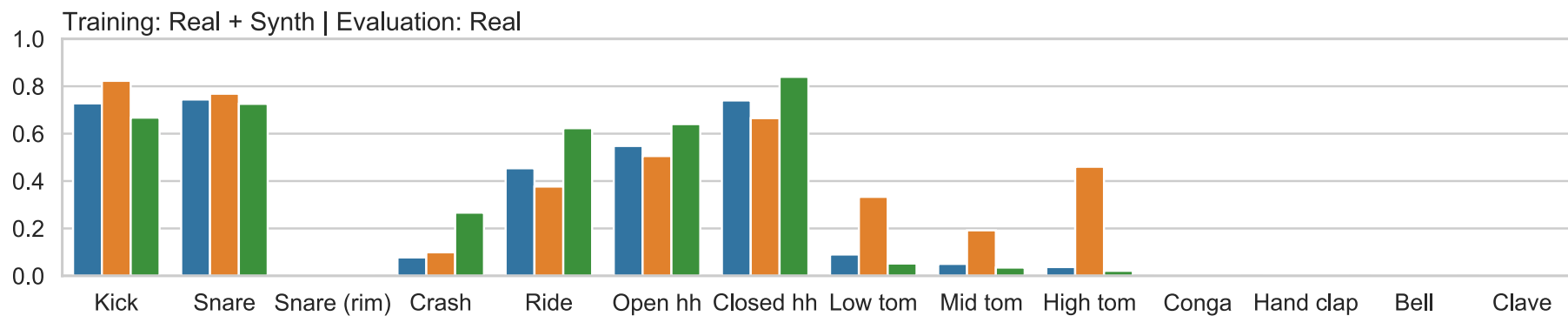
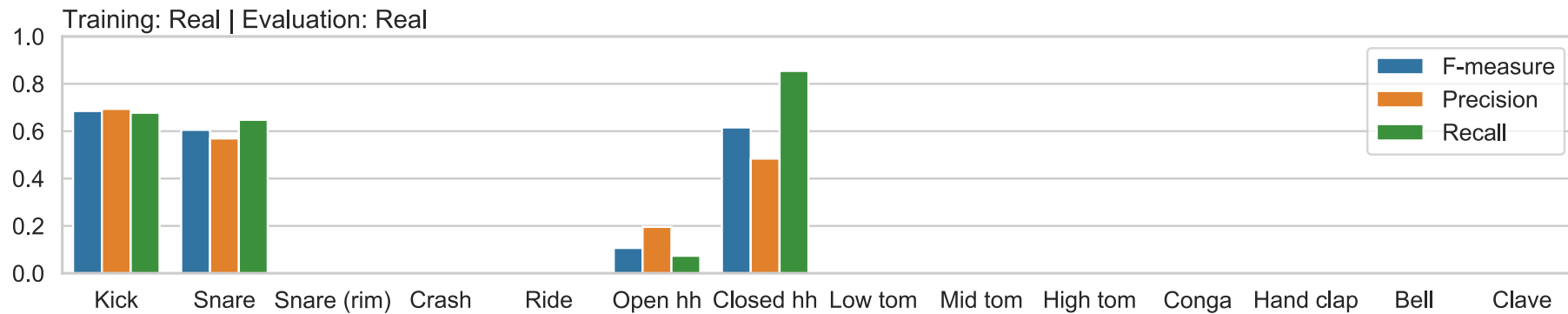
Small /
Single-task



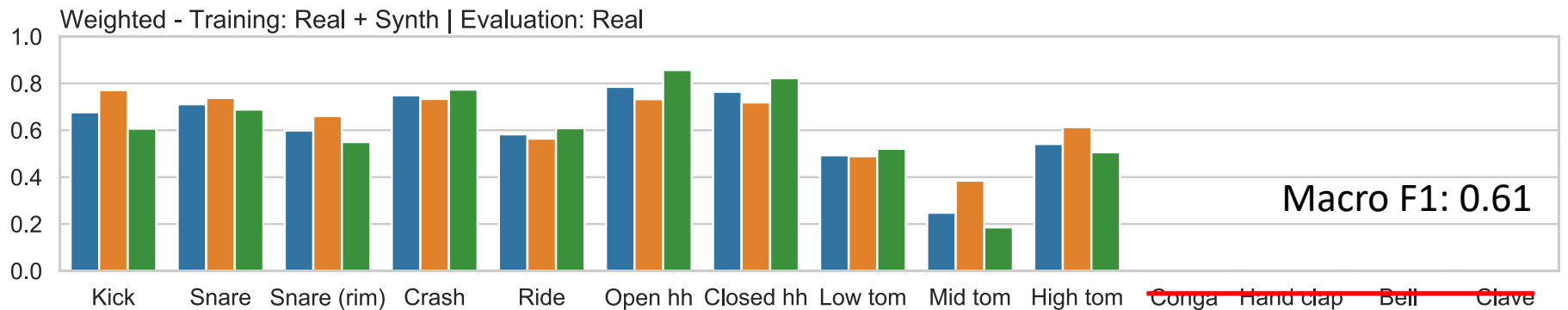
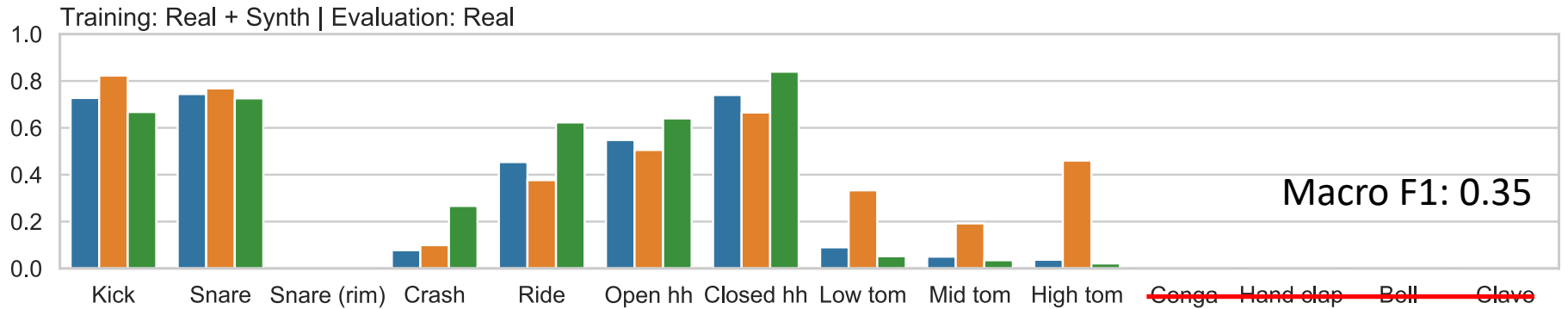
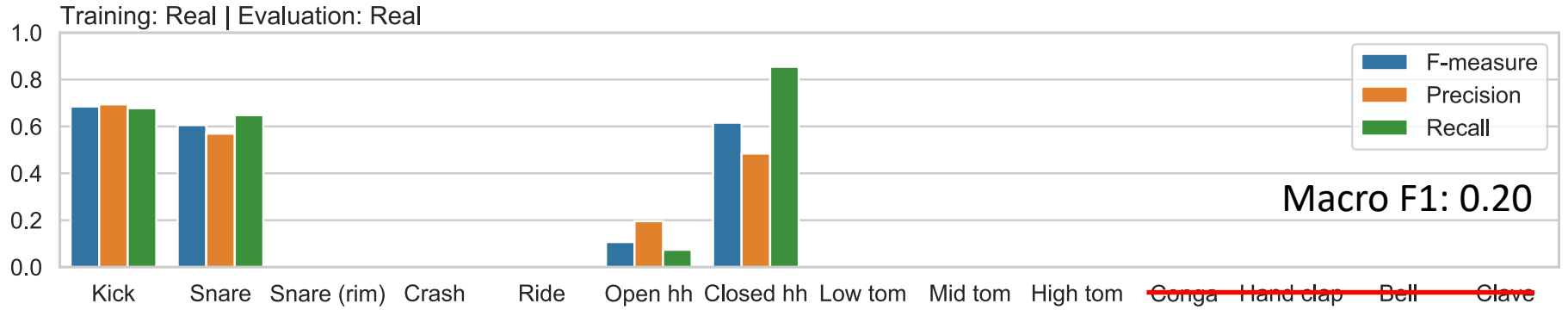
Results



Results

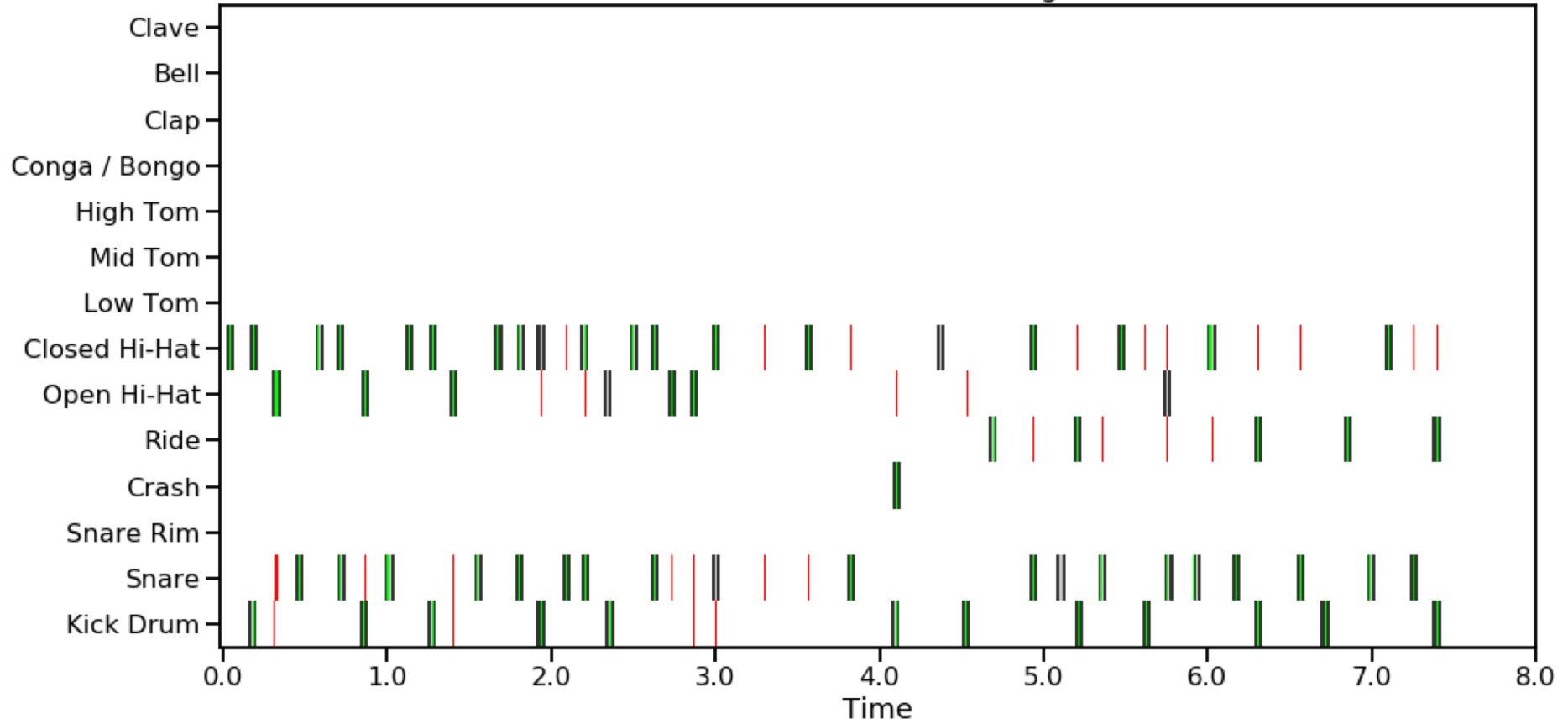


Results



Results

Small / Multi-task / Class-weighted



Conclusion

- Lots of mediocre synthetic data can improve performance on both overall performance and uncommon classes
- However, it must be used in conjunction with real music data
- Multi-task learning doesn't seem to help for large-vocab transcription, but does help in the auxiliary task of downbeat/beat tracking

**Download the trained models at
https://github.com/mcartwright/dafx2018_adt**