NISA (Non-Intrusive Signal Analysis) Development, Release, Noise Type Classifier

Lucia Eve Berger

M. Sci in Software Engineering

Managers:

DoYeong Kim, Dushyant Sharma

Outline



Objectives



Pipeline

NISA++

Feature extractions

Model Selection



Results



Integration & Release

Objectives

Expand the NISA++ development, using alternative feature extraction and smaller architectures

- Experiment with CNN architectures
- More flexible data provision and feature extraction

High Level

Refactor the NISA framework

- More flexible training and testing {controlled via config}
- Exploited latest CNN technology and feature extraction

NISA++ Release

- Released 16K, 8K model
- Released the SwishNet VAD for general use

Developed Model for Noise Type Classification

• Classification model for the Noise Type/Codec Model

NISA Pipeline



Preprocessing

Selection & corruption

Pipeline



Training

Supervised problem



Testing

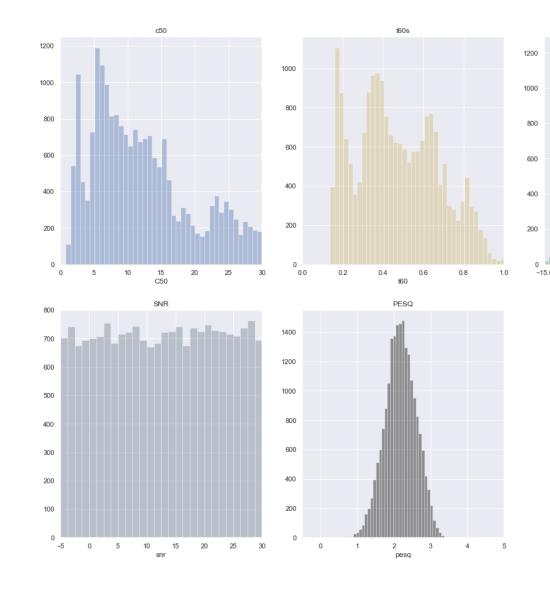
Comparing against larger baselines

DRR

Training NISA

- Wallstreet journal corpus
- Sampled large range of metrics
 - Sampled within bins
 - Augmentation
- ~20,038 utterances

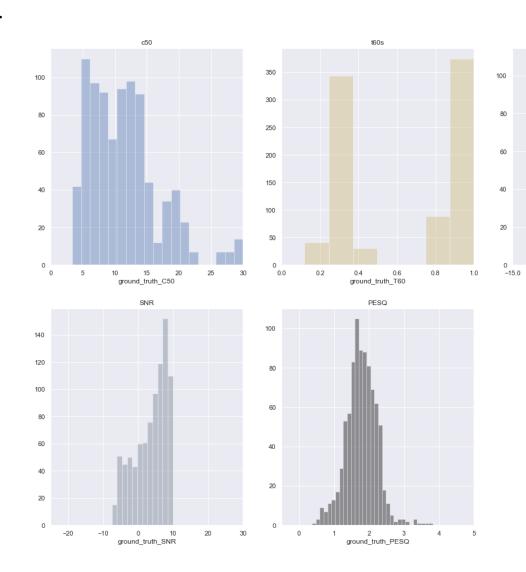
Paramet	Definition
er	
C50	Speech clarity
T60	Estimators of room
	reverberation time
SNR	Signal to noise ratio
DRR	Direct-to-reverberation-
	ratio
PESQ	Perceptual Evaluation of
	Speech Quality
VAD	Voice activity detection



ground truth DRR

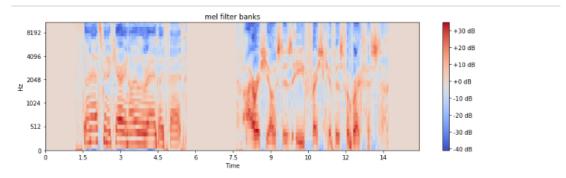
Testing Data {ASYM}

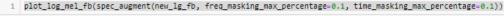
- Corrupted with corruption toolkit
- Highly noisy and reverberant
- Subset of libre-test data

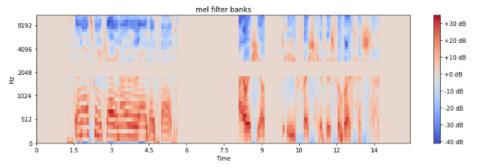


Feature Extractions

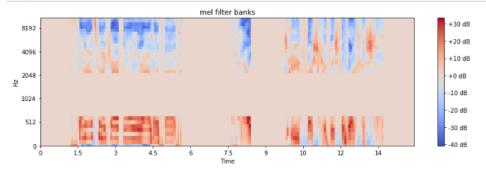
- Log Mel Filter-banks (40/80)
 - Spec Augmentation
 - Different frequency components
- PASE
 - Problem agnostic speech features
 - waveform based with out of box, no retraining
 - "derive useful speech representations by employing a self-supervised encoderdiscriminator approach"
 - https://arxiv.org/abs/1904.03416
- MREC features
 - LMFCCs, MDCCs, FFVs





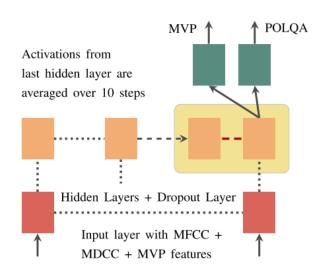






Model Selection 1

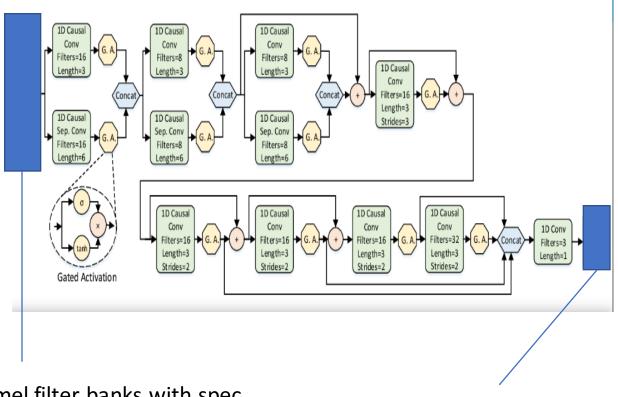
- Robust frame-wise multi-parameter estimator
- Experimented several Neural Networks
 - 3-layer Stacked LSTM:
 - n_hidden1=80/100
 - n_hidden2=54
 - n_hidden3=27
 - With output: {averaged over 10 steps}
- Advantage to this strategy: Given that LSTMs operate on sequence data, it means that the addition of layers adds levels of abstraction of input observations over time. In effect, chunking observations over time or representing the problem at different time scales.
 https://machinelearningmastery.com/stacked-long-short-term-memory-networks/



Model Selection 2

• 1D CNN

- Model that we use for the VAD binary classification problem
- Modified
 - Output layer
 - Number of filters
 - Length of kernals
 - Optimization parameters
 - https://arxiv.org/abs/1812.00149



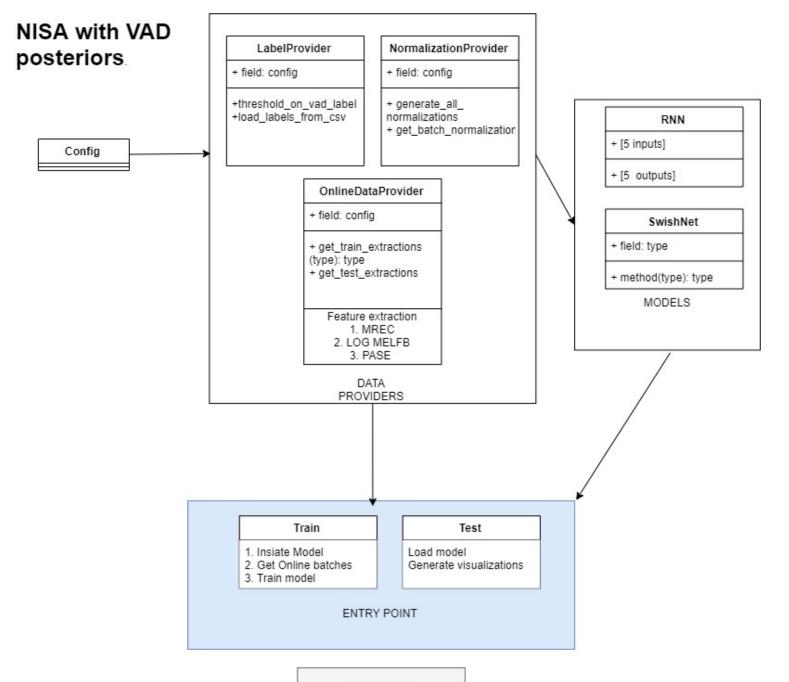
80 log mel filter banks with spec

Linear output with 5 nodes

Training and Testing Architecture

- Config controls the Training and testing outside of the entry point
- More flexible
 - Extraction type
 - Model Type {LSTM} or {SWISH}
 - Modularized/Customizable outside of the codebase

```
[INPUT DATA]
save dir={directory where the model is saved}
normalization parameters=normalization file: if this does not exist, that it will be created
logger file=
train data=OFFLINE EXTRACTED FEATURE DIRECTORY
train vad labels=GROUND TRUTH VAD LABELS/
train score labels= csv of train ground truth.csv
uses vad posteriors=0: FALSE
use vad features=0: FALSE
uses vad ground truth=1:
validation data=not used anymore
extraction type=mrec * very important
features input size=268 *very important: sets the size of the input on the y axis
file ending=*.txt *feature extraction type{file ending type}
window size= Size of the chunk
snr labels= extracted SNR labels
uses snr labels= whether the extractor SNR labels are used {Boolean}
[MODEL PARAMETERS]
model type=LSTM or SWISH: how the training parses which model to load
train batch size= the size of the batches per training
train num epochs= epochs per training
number_of_indices= count of chunks that are taken out each time
dropout keep prob=1.0
learning rate= starting learning rate
last filter size= {only for the SWISH}
input nodes=6 = {how many parameters to estimate}
starting_dropout=0.7 {starting dropout}
uses weighting=1 {whether to use the weighting vectors}
```



Visualization

Evaluation

Evaluation Explanation

- Models:
- LSTM: Stacked 3 layer LSTM that has three hidden layers. Each of these LSTM layers contains multiple memory cells.
- **CNN:** 1D CNN with larger filters than SwishNet Model: *Across temporal domain feature extractions as channels*

- RMSE: Root Mean Squared Error
- MAE: Mean Absolute Error
- Correlation: Pearson Correlation between Ground Truth and Predicted

ASYM Test Results

- Internal Dataset
- Same window size (350 ms)

PA R A M E T E R S	LARGER {FILTERS} CNN	LSTM + LOG MEL FILTERBANKS	BASELINE LSTM + MR E C
C50 RMSE	4.05	3.5	3.45
C50 correlation	0.756	0.747	0.84
T60 RMSE	0.344	0.3	0.35
t60 correlation	0.67	0.63	0.785
DRR RMSE	2.44	2.71	5.66
DRR correlation	0.39	0.369	0.562
Snr RMSE	3.744	3.54	4.02
snr correlation	0.75	0.7	0.9369
PESQ RMSE	0.28	0.28	0.26
PESQ correlation	0.83	0.85	0.855
VAD	0.932	0.929	
Trainable Parameters:	~54,000	~89,000	~124,000

Narrowband NISA Release ASYM data

- To be used with the NTE/VM2T
 - products
 - Created 8KHz SCT configs from 16K

setup

--sample_rate=8000

ASYM: Value	RMSE	MAE	Pearson
C50	3.46	2.66	0.79
Т60	0.295	0.23	0.75
SNR	3.54	2.4	0.774
DRR	2.65	2.03	0.4
PESQ	0.29	0.23	0.866
VAD	F1score: 0.925		

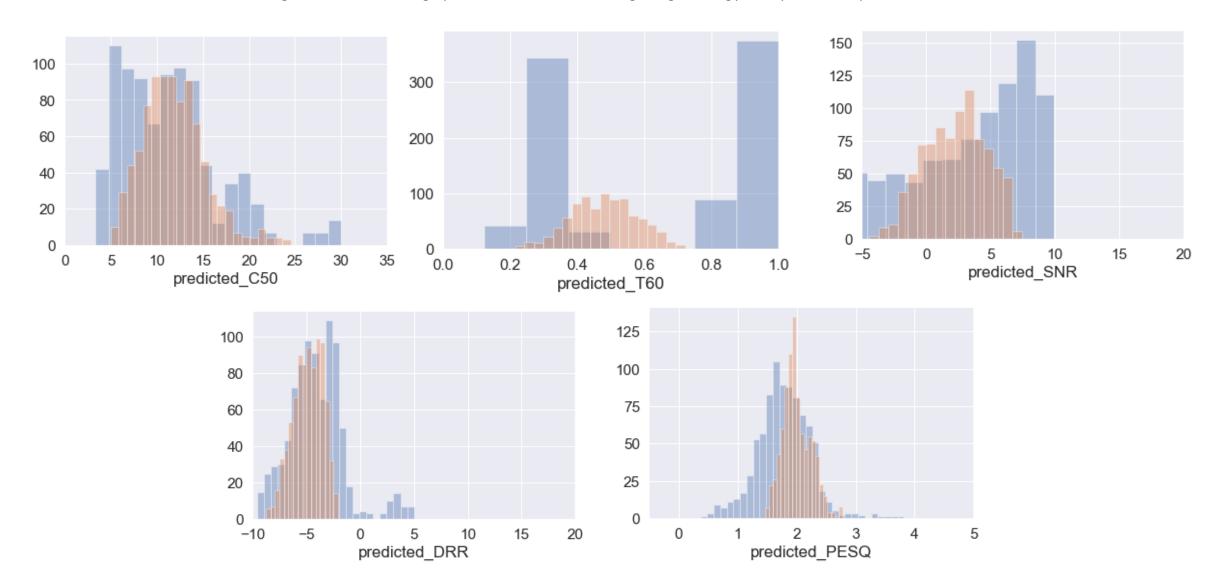
ACE Test Data-Set for the EUSPICO paper

- Published dataset
- Focused on Reverb parameters

METRICS	M F B + S A + C N N	MFB+SA+ LSTM	MFB+ LSTM	PASE+ LSTM	MREC + LSTM
RMSE C50	3.5	2.922	2.99	4.58	3.205
MAE C50	2.9	2.35	2.38	3.71	2.5185
RMSE SNR	3.8	4.3	5.3	3.94	3.75
MAE SNR	3.17	3.5	4.26	3.1	2.97
TRAINABLE PARAMETE RS:	16,346	125824	125824	125824	125824

Evaluation Distributions: Missing outliners?

Blue ground truth, Orange predicted, add some weighting strategy to capture full spread!



Real Time Factor Estimate

NISA Parameters Estimate	CNN MELFB + Spec:	LSTM + MELFB + SPEC
10 samples {per utterance}	~0.01 s	~0.076 s

Model Loading	CNN MELFB + Spec:	LSTM + MELFB + SPEC
One Time Cost	~1.31s	4.45

Observations:

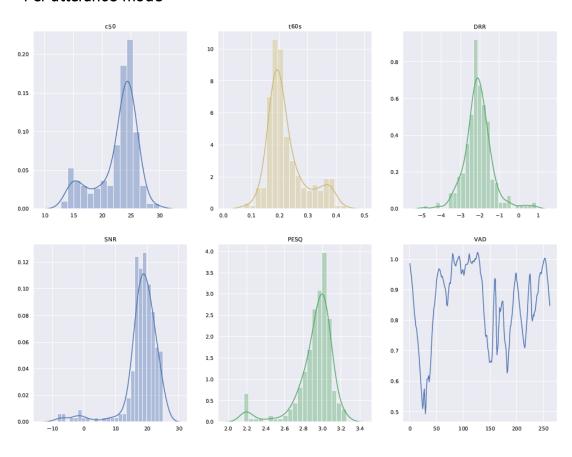
Very little change in RMSE with 10 samples or iterating by 1 (increase by x5)

- Python environment: very slow
- Hardware variability:
- time estimates.xlsx

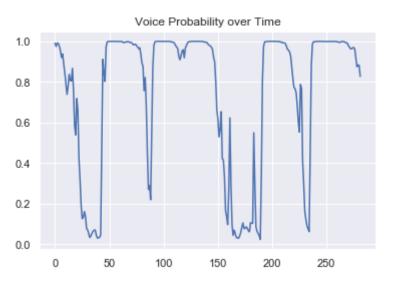
Release of NISA

NISA++ per utterance mode

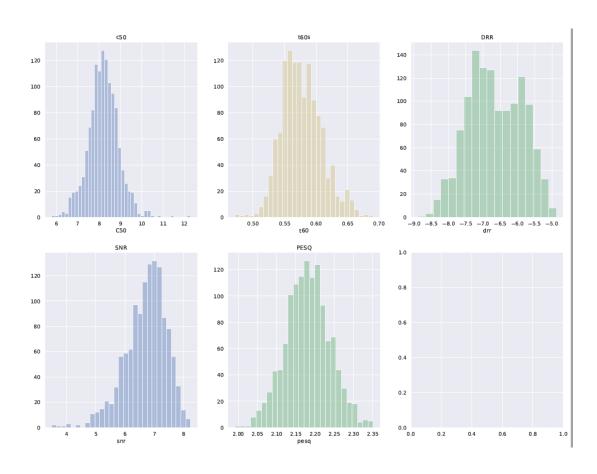
Per utterance mode

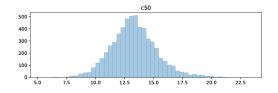


SwishNet VAD estimator



NISA ++ Batch Mode





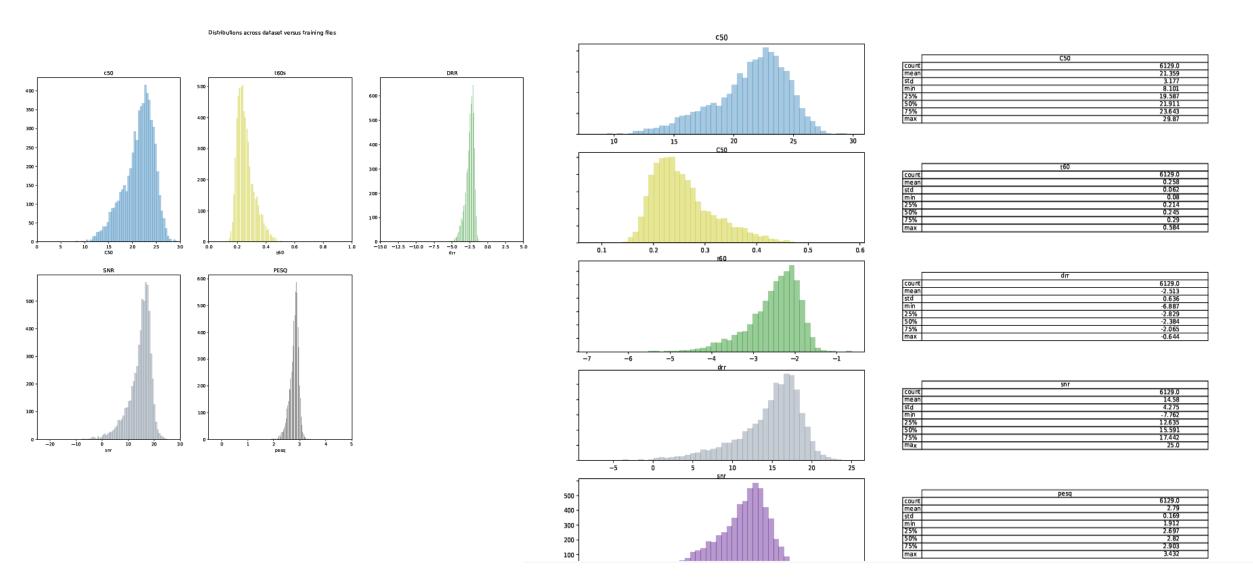


- 1) High Level Distributions
- 2) Min/Max, Quartile information per parameter
- 3) SwishNet VAD posteriors

```
/gpfs/amr_alg/others/lucia/conda/envs/TF_GPU/bin/python run_NISA.py
--is_batch=[0,1] : whether the code is run as a single file or many files
--file_path={}: file or list of files
--output_dir={}: where the output will be written
--is_verbose=1: whether to output pdfs as visualization tools
--vad_threshold=0-1: threshold for the vad posterior
--run_vad_only=0: whether to run only SwishNet

OPTIONAL PARAMS:
--is_verbose_per_utterance=1: whether to output per file pdfs
--sample rate: desired sample rate {possibilities are 8000,16000}
```

NTE Charter Use-Case

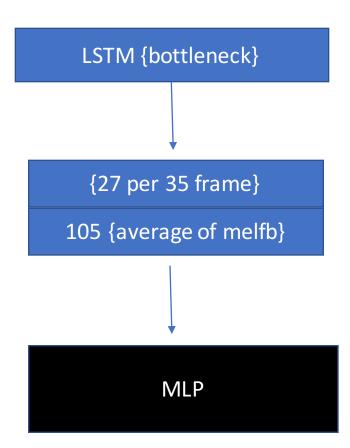


Noise Type Classifier/CODEC

Ongoing

Noise Type Classifier

- Noise Types:
 - Ambient, Babble, Music, White
- 2 Approaches:
 - Baseline Swishnet (1D CNN)
 - Bottleneck LSTM => MLP



Evaluation

SwishNet Classification:

1 second window size

MLP Bottleneck:

1.05 second window size

	Predicted AMBIENT	Predicted BABBLE	Predicted MUSIC	Predicted WHITE
Ground Truth AMBIENT	0.83	0.13	0.027	0.001
Ground Truth Babble	0.22	0.67	0.13	0.04
Ground Truth MUSIC	0.16	0.10	0.69	0.07
GROUND TRUTH WHITE	0.02	0.002	0.01	0.97

	Predicted AMBIENT	Predicted BABBLE	Predicted MUSIC	Predicted WHITE
Ground Truth AMBIENT	0.46	0.37	0.1	0.05
Ground Truth Babble	0.16	0.79	0.021	0.021
Ground Truth MUSIC	0.44	0.47	0.05	0.03
GROUND TRUTH WHITE	0.202	0.024	0.006	0.76

Take-Aways & Next Steps

Hands-on exposure to audio data processing, corruption and modelling

- 1. Hand-off to Cheng ©
- 2. Submitted a Paper to EUSIPCO 2020
 - Non-Intrusive Estimation of Speech Signal Parameters using a Frame-based Machine Learning Approach

Questions?

Thank you!

Special thanks to Dushyant & Carl for the mentorship.

Code:

NISA: https://git.labs.nuance.com/dushyant.sharma/NISA

Literature Reviews: https://nuance.jiveon.com/people/lucia.berger/blog

Sources: https://jeddy92.github.io/JEddy92.github.io/ts_seq2seq_conv/