Detecting anxiety from short clips of free-form speech

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Motivation

- Mental wellbeing is quintessential for overall health but many social issues like stigma and monetary barriers exist.
- Mental assessments are intermittent and may be limited partly due to the episodic nature of psychiatric symptoms.
- The gravity of the problem combined with the massive scope of improving upon current methods motivates us to study the problem of detecting anxiety disorders through speech.

Problem Statement

- Given the raw audio recording of the user, predict if the user has anxiety disorder or not as measured by the GAD-7 [4] questionnaire.
- Score of 5 or greater on GAD-7 implies some form of anxiety

Over the <u>last 2 weeks</u> , how often have you been bothered by the following problems? (Use "\sum " to indicate your answer)	Not at all	Several days	More than half the days	Nearl every o
Feeling nervous, anxious or on edge	0	1	2	3
2. Not being able to stop or control worrying	0	1	2	3
3. Worrying too much about different things	0	1	2	3
4. Trouble relaxing	0	1	2	3
5. Being so restless that it is hard to sit still	0	1	2	3
6. Becoming easily annoyed or irritable	0	1	2	3
Feeling afraid as if something awful might happen	0	1	2	3

Data

- Generalized Anxiety Disorder (GAD) dataset from Kintsugi Mindful Wellness, Inc.
- Self audio journal entry from participating users.
- Labels obtained by scores of GAD-7 questionnaire filled by the users shortly after the recorded journal.
- 2257 labelled examples with raw audio and test scores

Methods

- Audio Features (GeMAPS)
- Transcript based classifier
- Wav2Vec embeddings
- Multi-modal anxiety detection

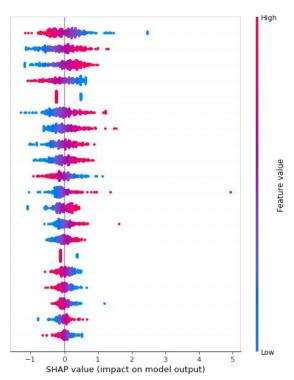
Audio Features (GeMAPS)

- Standard acoustic parameter set for various areas of automatic voice analysis including clinical speech analysis [3]
 - Jitter/Shimmer
 - Loudness
 - Pitch (F0)
- Transcript based features
 - Emotion
 - Sentiment
- Logistic regression model with L1 regularization

Model	Precision	Recall	F1	AUROC
Random	0.50	0.48	0.49	0.50
GeMAPS features	0.63	0.53	0.58	0.66

Audio Features (GeMAPS)

F0semitoneFrom27.5Hz_sma3nz_amean loudness_sma3_stddevFallingSlope F1amplitudeLogRelF0 sma3nz amean F0semitoneFrom27.5Hz sma3nz stddevRisingSlope sentiment_positive F2amplitudeLogRelF0_sma3nz_stddevNorm F0semitoneFrom27.5Hz_sma3nz_meanRisingSlope HNRdBACF sma3nz amean mfcc3_sma3_amean mfcc1_sma3_amean spectralFlux sma3 stddevNorm F0semitoneFrom27.5Hz_sma3nz_percentile20.0 hammarbergIndexUV sma3nz amean mfcc4V sma3nz amean emotion_joy loudness sma3 stddevRisingSlope mfcc2V sma3nz amean alphaRatioUV_sma3nz_amean F0semitoneFrom27.5Hz sma3nz percentile80.0 MeanVoicedSegmentLengthSec



Transcript based classifier

 In this approach, we don't deal with audio data directly, but instead have a 2-step process:

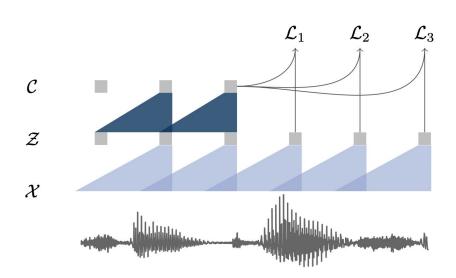
 In the first step, we use Wav2Vec2-Base model to transcribe the given audio files. We then use pre-trained Roberta Large optimized for Semantic Textual Similarity (STS) to generate sentence-level embeddings for the transcripts.

 In the second step, the generated sentence embeddings and their corresponding Anxiety/No Anxiety labels are used to train a GBC model to serve as a binary classifier.

Model	Precision	Recall	F1	AUROC
Random	0.50	0.48	0.49	0.50
GeMAPS features	0.63	0.53	0.58	0.66
Transcript classifier	0.64	0.57	0.60	0.68

Wav2Vec Embedding

- Embed the raw audio input into a sequence of low-dimensional embeddings using wav2vec [5]
- Train a SVM classifier over the mean of the sequence of embeddings



Model	Precision	Recall	F1	AUROC
Random	0.50	0.48	0.49	0.50
GeMAPS features	0.63	0.53	0.58	0.66
Transcript classifier	0.64	0.57	0.60	0.68
Wav2Vec features	0.66	0.61	0.64	0.69

Joint Audio and Text Model

- Final model, combines audio and text features in a multi-modal fashion. [Siriwardhana, Shamane, et al]
- Raw text transcripts tokenized and fed to a pre-trained Roberta model to generate a text embedding(CLS_{text}).
- A VQ-Wav2Vec model used to create discretized representation of raw audio, fed into a pre-trained BERT to generate CLS_{speech}.
- Finally a linear layer, followed by softmax applied on concatenated embeddings for label prediction.
- Currently working on this model implementation.

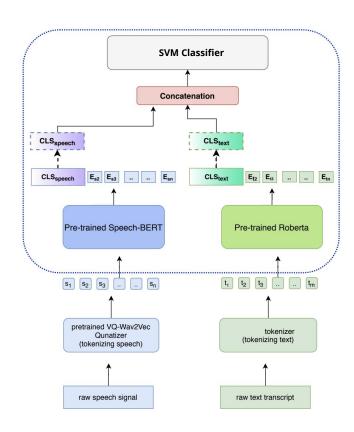


Image and model from here.

Model	Precision	Recall	F1	AUROC
Random	0.50	0.48	0.49	0.50
GeMAPS features	0.63	0.53	0.58	0.66
Transcript classifier	0.64	0.57	0.60	0.68
Wav2Vec features	0.66	0.61	0.64	0.69
Multi modal classifier	0.66	0.60	0.61	0.68

Future work

- Use GAD-7 based sample weights to see if the model improves
- Also aim to qualitatively analyze and compare the performance of different models

References

- 1. Siriwardhana, Shamane, et al. "Jointly Fine-Tuning" BERT-like" Self Supervised Models to Improve Multimodal Speech Emotion Recognition." *arXiv preprint arXiv:2008.06682* (2020).
- 2. Baevski, Alexei, et al. "wav2vec 2.0: A framework for self-supervised learning of speech representations." *arXiv preprint arXiv:2006.11477* (2020).
- 3. Eyben, Florian, et al. "The Geneva minimalistic acoustic parameter set (GeMAPS) for voice research and affective computing." IEEE transactions on affective computing 7.2 (2015): 190-202.
- 4. Spitzer, Robert L., et al. "A brief measure for assessing generalized anxiety disorder: the GAD-7." *Archives of internal medicine* 166.10 (2006): 1092-1097.
- 5. Schneider, Steffen, et al. "wav2vec: Unsupervised pre-training for speech recognition." arXiv preprint arXiv:1904.05862 (2019).