Crowdsourcing tool for annotated speech corpora creation

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Outline

1. Introduction

2. Broadcast speech annotation

3. Correcting annotations for crowdsourced audio

«Classical» approach:

- Acoustic models => phonemes
- Pronunciation model => words
- Language model => phrase

End-to-end approaches:

- Better accuracy, require a large amount of training data
- Using data for related languages; pre-trained models

Wav2vec2:

- NLP/Computer Vision fields benefited from using self-supervised pretraining
- Allows to learn **robust** audio representations based **on unlabeled data**
- Masking fragments; model tries to distinguish the true speech representation from distractors (uniformly sampled from other masked fragments)

The main benefits:

- For low-resourced language it's much easier to find unlabeled data
- ASR system becomes more robust to background noises, dialects, speakers

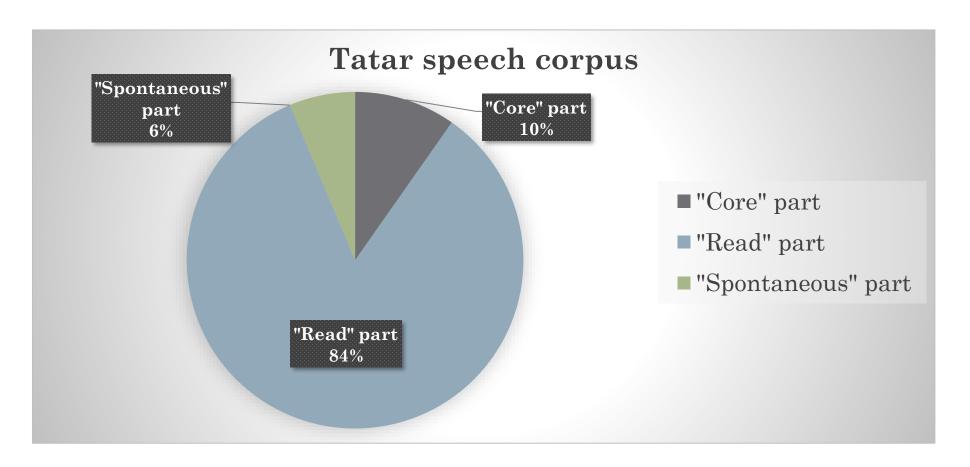
Main goals:

- Collect required unlabeled and labeled Tatar speech corpora
- Try the approach with iterative self-supervised pretraining steps on audio data that is increasingly closer to the target domain

• Recordings' format: 16 kHz, 16 bps mono WAV PCM

Speakers: native speakers, Kazan dialect

• Speech type: read speech



Core part

- Manually collected separate words and phrases
- Phonetically full, max context
- 251 speaker, average duration 0:01:58
- Total duration 8:12:16

• Read part:

- Rule-based selection from text corpus
- 190 speakers, average duration 0:22:18
- Total duration 70:39:00

Spontaneous part:

- Non-overlapping dialogues
- Total duration -5:19:33

| Speech corpus | | | |
|------------------------|-----------|--|--|
| # speakers | 499 | | |
| Duration | 99:09:59 | | |
| Male / Female | 30% / 70% | | |
| $Spontaneous\ speech*$ | 5:19:33 | | |

^{*} We're recording spontaneous speech too, but it's not annotated

Annotation

- Speaker's name
- Age
- Gender
- Native language
- Nationality
- Speech quality (expert's mark from 1 to 5)
- Dialect
- Microphone model
- Comment

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Main goal – tools for corpus creation.

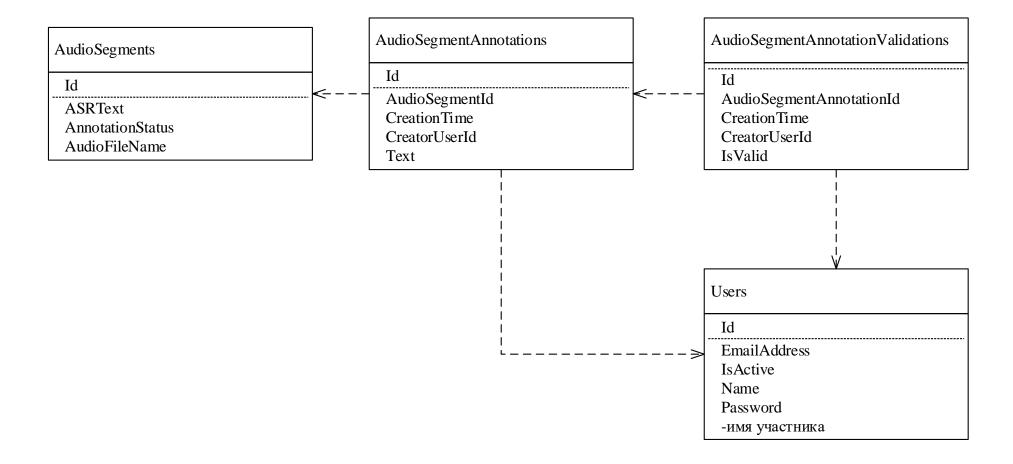
1. Broadcast speech annotation web-tools

[self-supervised approaches]

2. Tool to check and correct annotations

- ASP.Net Core
- React.js
- DDD (Domain Driven Design):
 - Infrastructure Layer
 - Domain layer
 - Application Layer
 - Service Layer
 - Presentation Layer
 - Client Applications

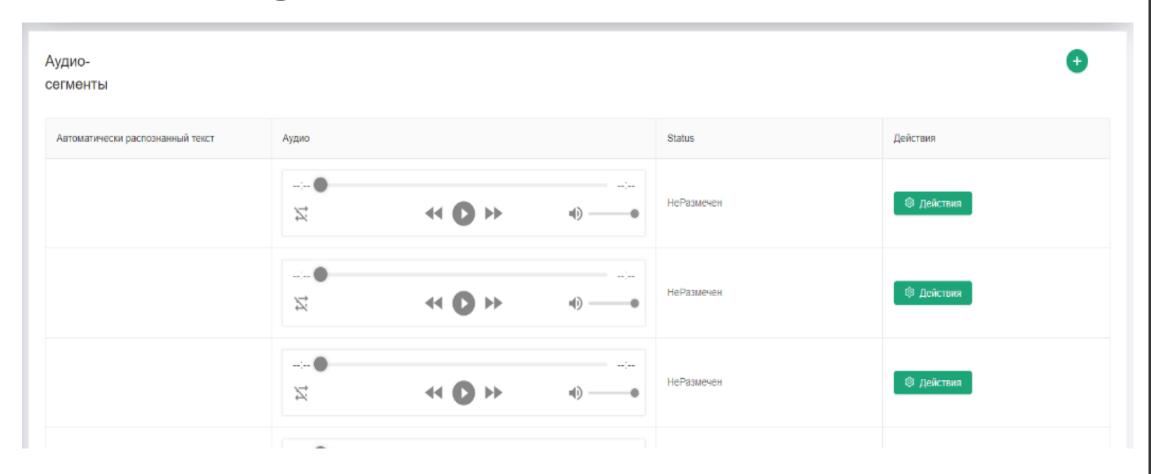
2. Project description PostgreSQL



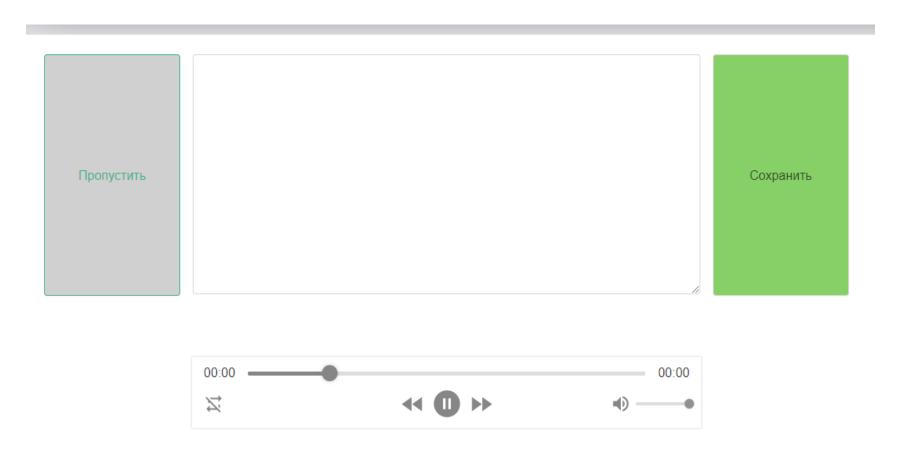
Basic functionality:

- Audio files upload;
- VAD and splitting uploaded files into fragments;
- Web-form for annotating fragment;
- Web-form for validating made annotations;
- View status of annotation of all segments;
- Downloading the annotations.

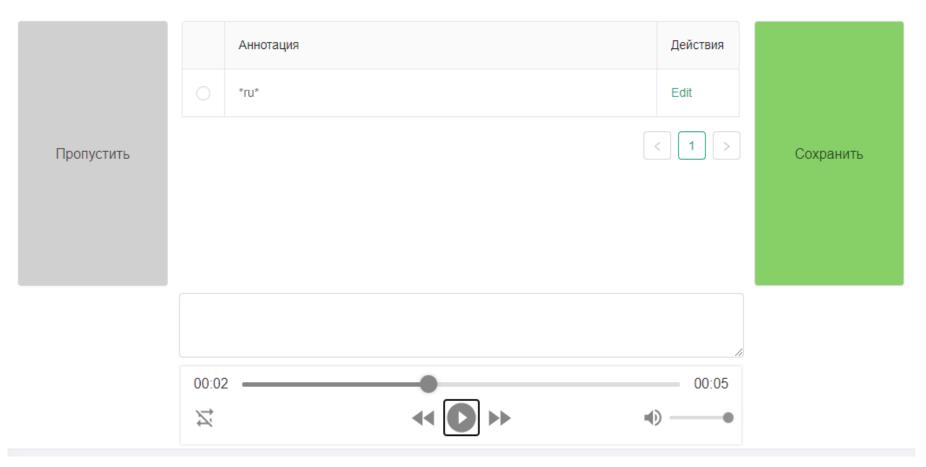
View fragments' statuses



Annotating fragments



Validating fragments



Initial data:

- From TNV Planeta broadcast company;
- Recordings from December 2019;
- AVI video with mp3 96 kB/s stereo audio signal;
- Converted to 16 bps 16 kHz WAV;
- Total duration 733 hour.

We manually selected segments for the first stage annotation:

- News programs;
- Interviews;
- Talk-shows.

In total 40 segments (23 hours 21 minutes) have been uploaded to the system.

This gave us 22 432 audio fragments with a duration less than 15 seconds.

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3. Telegram bot

@TatarVoiceBot

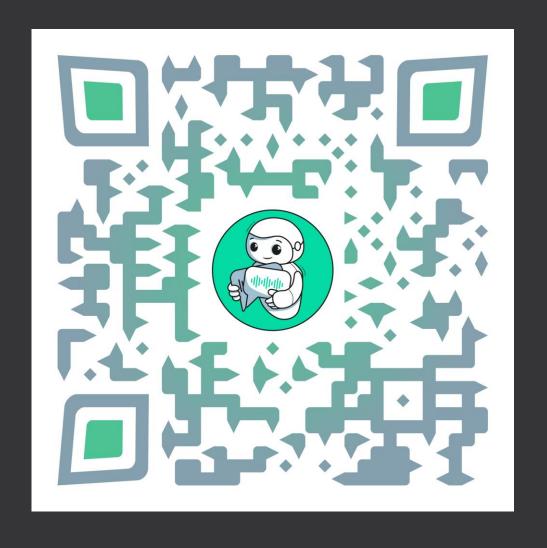
Goal – 500 hours

- 408 speakers
- 9 hours 28 minutes

Commands:

- Next new phrase to read and send as voice message;
- 2. Correct record previous phrase again;
- 3. Skip to skip current phrase;
- 4. Statistics show user's and overall statistics;
- 5. Age select age interval;
- 6. Russian/Tatar/Help.





@TatarVoiceBot

3. Process

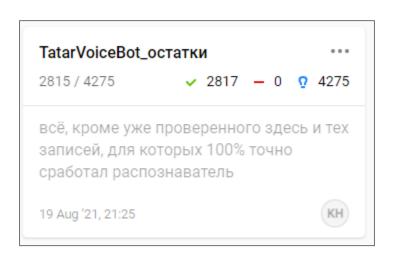
People read phrases (12 000)

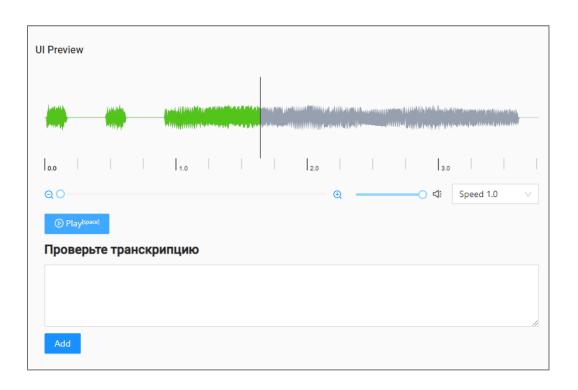
ASR recognizes text

Manual check for files with errors (4 200)

https://github.com/heartexlabs/label-studio:

- Allows to build universal platform for all Institute annotation tasks;
- Easy to configure for each task (interface, data, hotkeys, etc.);
- Local, free.





Requirements:

- Unlabeled dataset for SS steps;
- Labeled dataset for FT steps.

Labeled datasets:

- 1. «Tatar Corpus»;
- 2. «Common Voice»;
- 3. [new] TV broadcasting;
- 4. [new] TatarVoiceBot.

Unlabeled dataset:

- 1. Audiobooks (read speech, recording studio) -114 hours;
- 2. TV broadcasting recordings for 1 month (spontaneous speech, bg noises, music) 733 hours;
- 3. 2 radio stations archives (read and spontaneous, bg music) -215 hours;
- 4. 100 scientific lectures from YouTube (good SNR) 87 hours.

Preprocessing:

- 1. Audio track extraction;
- 2. Audio conversion to 16 bps 16 kHz mono format;
- 3. VAD;
- 4. Filtering short (<4.5 sec) and long (>30 sec) fragments.

Statistics of final unlabeled dataset

| Subcorpus | Initial | After VAD | After filtering short and long fragments |
|----------------|-----------------|---------------------|------------------------------------------|
| Audiobooks | 114 hours | 105 hours | 58 hours |
| | (520 fragments) | (36 712 fragments) | (17 563 fragments) |
| TV | 733 hours | 472 hours | 202 hours |
| | (62 fragments) | (263 466 fragments) | (67 065 fragments) |
| Radio stations | 215 hours | 146 hours | 29 hours |
| | (398 fragments) | (29 778 fragments) | (8 941 fragments) |
| YouTube videos | 87 hours | 81 hours | 39 hours |
| | (100 fragments) | (31 437 fragments) | (12 764 fragments) |
| Corpus | 1 150 hours | 804 hours | 328 hours |

Thank you

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