Disvoice Documentation

Release 0.1

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DisVoice is a python framework designed to compute features from speech files. Disvoice computes glottal, phonation, articulation, prosody, and phonological-based features both from sustained vowels and continuous speech utterances with the aim to recognize praliguistic aspects from speech. The features can be used in classifiers to recognize emotions, or communication capabilities of patients with different speech disorders including diseases with functional origin such as larinx cancer or nodules; craneo-facial based disorders such as hipernasality developed by cleft-lip and palate; or neurodegenerative disorders such as Parkinson's or Hungtinton's diseases.

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GLOTTAL FEATURES

class glottal.Glottal

Compute features based on the glottal source reconstruction from sustained vowels and continuous speech.

For continuous speech, the features are computed over voiced segments

Nine descriptors are computed:

- 1. Variability of time between consecutive glottal closure instants (GCI)
- 2. Average opening quotient (OQ) for consecutive glottal cycles-> rate of opening phase duration / duration of glottal cycle
- 3. Variability of opening quotient (OQ) for consecutive glottal cycles-> rate of opening phase duration /duration of glottal cycle
- 4. Average normalized amplitude quotient (NAQ) for consecutive glottal cycles-> ratio of the amplitude quotient and the duration of the glottal cycle
- 5. Variability of normalized amplitude quotient (NAQ) for consecutive glottal cycles-> ratio of the amplitude quotient and the duration of the glottal cycle
- 6. Average H1H2: Difference between the first two harmonics of the glottal flow signal
- 7. Variability H1H2: Difference between the first two harmonics of the glottal flow signal
- 8. Average of Harmonic richness factor (HRF): ratio of the sum of the harmonics amplitude and the amplitude of the fundamental frequency
- 9. Variability of HRF

Static or dynamic matrices can be computed:

Static matrix is formed with 36 features formed with (9 descriptors) x (4 functionals: mean, std, skewness, kurtosis)

Dynamic matrix is formed with the 9 descriptors computed for frames of 200 ms length with a time-shift of 100 ms

Notes:

1. The fundamental frequency is computed using the RAPT algorithm.

Examples command line:

Examples directly in Python

```
\textbf{extract\_features\_file} \ (\textit{audio}, \textit{static=True}, \textit{plots=False}, \textit{fmt='npy'}, \textit{kaldi\_file=''})
```

Extract the glottal features from an audio file

Parameters

- audio .wav audio file.
- **static** whether to compute and return statistic functionals over the feature matrix, or return the feature matrix computed over frames
- plots timeshift to extract the features
- **fmt** format to return the features (npy, dataframe, torch, kaldi)
- kaldi_file file to store kaldi features, only valid when fmt=="kaldi"

Returns features computed from the audio file.

extract_features_path (path_audio, static=True, plots=False, fmt='npy', kaldi_file='') Extract the glottal features for audios inside a path

Parameters

- path_audio directory with (.wav) audio files inside, sampled at 16 kHz
- **static** whether to compute and return statistic functionals over the feature matrix, or return the feature matrix computed over frames
- plots timeshift to extract the features
- **fmt** format to return the features (npy, dataframe, torch, kaldi)
- kaldi_file file to store kaldifeatures, only valid when fmt=="kaldi"

Returns features computed from the audio file.

plot_glottal (data_audio, fs, GCI, glottal_flow, glottal_sig, GCI_avg, GCI_std)
Plots of the glottal features

Parameters

- data_audio speech signal.
- **fs** sampling frequency
- GCI glottal closure instants
- glottal_flow glottal flow
- $\bullet \ \, \textbf{glottal_sig} reconstructed \ glottal \ signal \\$
- GCI_avg average of the glottal closure instants
- **GCI_std** standard deviation of the glottal closure instants

Returns plots of the glottal features.

CHAPTER

TWO

PHONATION FEATURES

Created on Jul 21 2017

@author: J. C. Vasquez-Correa

class phonation. Phonation

Compute phonation features from sustained vowels and continuous speech.

For continuous speech, the features are computed over voiced segments

Seven descriptors are computed:

- 1. First derivative of the fundamental Frequency
- 2. Second derivative of the fundamental Frequency
- 3. Jitter
- 4. Shimmer
- 5. Amplitude perturbation quotient
- 6. Pitch perturbation quotient
- 7. Logaritmic Energy

Static or dynamic matrices can be computed:

Static matrix is formed with 29 features formed with (seven descriptors) x (4 functionals: mean, std, skewness, kurtosis) + degree of Unvoiced

Dynamic matrix is formed with the seven descriptors computed for frames of 40 ms.

Notes:

- 1. In dynamic features the first 11 frames of each recording are not considered to be able to stack the APQ and PPQ descriptors with the remaining ones.
- 2. The fundamental frequency is computed the RAPT algorithm. To use the PRAAT method, change the "self.pitch method" variable in the class constructor.

Script is called as follows

Examples command line:

Examples directly in Python

 $\textbf{extract_features_file} \ (\textit{audio}, \textit{static=True}, \textit{plots=False}, \textit{fmt='npy'}, \textit{kaldi_file=''})$

Extract the phonation features from an audio file

Parameters

- audio .wav audio file.
- **static** whether to compute and return statistic functionals over the feature matrix, or return the feature matrix computed over frames
- plots timeshift to extract the features
- **fmt** format to return the features (npy, dataframe, torch, kaldi)
- kaldi_file file to store kaldi features, only valid when fmt=="kaldi"

Returns features computed from the audio file.

```
>>> phonation.extract_features_file(file_audio, static=False, plots=False, ofmt="kaldi", kaldi_file="./test")
```

extract_features_path (path_audio, static=True, plots=False, fmt='npy', kaldi_file='') Extract the phonation features for audios inside a path

Parameters

- path_audio directory with (.wav) audio files inside, sampled at 16 kHz
- **static** whether to compute and return statistic functionals over the feature matrix, or return the feature matrix computed over frames
- plots timeshift to extract the features
- **fmt** format to return the features (npy, dataframe, torch, kaldi)
- **kaldi file** file to store kaldifeatures, only valid when fmt=="kaldi"

Returns features computed from the audio file.

plot_phon (data_audio, fs, F0, logE)

Plots of the phonation features

Parameters

- data_audio speech signal.
- **fs** sampling frequency
- \bullet **F0** contour of the fundamental frequency
- logE contour of the log-energy

Returns plots of the phonation features.

THREE

ARTICULATION FEATURES

Created on Jul 21 2017

@author: J. C. Vasquez-Correa

class articulation. Articulation

Compute articulation features from continuous speech.

- 122 descriptors are computed:
- 1-22. Bark band energies in onset transitions (22 BBE).
- 23-34. Mel frequency cepstral coefficients in onset transitions (12 MFCC onset)
- 35-46. First derivative of the MFCCs in onset transitions (12 DMFCC onset)
- 47-58. Second derivative of the MFCCs in onset transitions (12 DDMFCC onset)
- 59-80. Bark band energies in offset transitions (22 BBE).
- 81-92. MFCCC in offset transitions (12 MFCC offset)
- 93-104. First derivative of the MFCCs in offset transitions (12 DMFCC offset)
- 105-116. Second derivative of the MFCCs in offset transitions (12 DMFCC offset)
- 117. First formant Frequency
- 118. First Derivative of the first formant frequency
- 119. Second Derivative of the first formant frequency
- 120. Second formant Frequency
- 121. First derivative of the Second formant Frequency
- 122. Second derivative of the Second formant Frequency

Static or dynamic matrices can be computed:

Static matrix is formed with 488 features formed with (122 descriptors) x (4 functionals: mean, std, skewness, kurtosis)

Dynamic matrix are formed with the 58 descriptors (22 BBEs, 12 MFCC, 12DMFCC, 12 DDMFCC) computed for frames of 40 ms with a time-shift of 20 ms in onset transitions.

The first two frames of each recording are not considered for dynamic analysis to be able to stack the derivatives of MFCCs

Notes: 1. The first two frames of each recording are not considered for dynamic analysis to be able to stack the derivatives of MFCCs 2. The fundamental frequency is computed the PRAAT algorithm. To use the RAPT method, change the "self.pitch method" variable in the class constructor.

Script is called as follows

```
>>> python articulation.py <file_or_folder_audio> <file_features> <static (true_ 

or false)> <plots (true or false)> <format (csv, txt, npy, kaldi, torch)>
```

Examples command line:

Examples directly in Python

extract_features_file (audio, static=True, plots=False, fmt='npy', kaldi_file='')

Extract the articulation features from an audio file

Parameters

- audio .wav audio file.
- **static** whether to compute and return statistic functionals over the feature matrix, or return the feature matrix computed over frames
- plots timeshift to extract the features
- fmt format to return the features (npy, dataframe, torch, kaldi)
- kaldi_file file to store kaldi features, only valid when fmt=="kaldi"

Returns features computed from the audio file.

extract_features_path (path_audio, static=True, plots=False, fmt='npy', kaldi_file='') Extract the articulation features for audios inside a path

Parameters

- path_audio directory with (.wav) audio files inside, sampled at 16 kHz
- **static** whether to compute and return statistic functionals over the feature matrix, or return the feature matrix computed over frames
- plots timeshift to extract the features
- **fmt** format to return the features (npy, dataframe, torch, kaldi)
- **kaldi file** file to store kaldifeatures, only valid when fmt=="kaldi"

Returns features computed from the audio file.

plot_art (data_audio, fs, F0, F1, F2, segmentsOn, segmentsOff)
Plots of the articulation features

Parameters

- data_audio speech signal.
- **fs** sampling frequency
- **FO** contour of the fundamental frequency
- **F1** contour of the 1st formant
- **F2** contour of the 2nd formant
- segmentsOn list with the onset segments
- segmentsOff list with the offset segments

Returns plots of the articulation features.

CHAPTER

FOUR

PROSODY FEATURES

Created on Jul 21 2017, Modified Apr 10 2018.

@author: J. C. Vasquez-Correa, T. Arias-Vergara, J. S. Guerrero

class prosody. Prosody

Compute prosody features from continuous speech based on duration, fundamental frequency and energy. Static or dynamic matrices can be computed: Static matrix is formed with 103 features and include

- 1-6 F0-contour: Avg., Std., Max., Min., Skewness, Kurtosis
- 7-12 Tilt of a linear estimation of F0 for each voiced segment: Avg., Std., Max., Min., Skewness, Kurtosis
- 13-18 MSE of a linear estimation of F0 for each voiced segment: Avg., Std., Max., Min., Skewness, Kurtosis
- 19-24 F0 on the first voiced segment: Avg., Std., Max., Min., Skewness, Kurtosis
- 25-30 F0 on the last voiced segment: Avg., Std., Max., Min., Skewness, Kurtosis
- 31-34 energy-contour for voiced segments: Avg., Std., Skewness, Kurtosis
- 35-38 Tilt of a linear estimation of energy contour for V segments: Avg., Std., Skewness, Kurtosis
- 39-42 MSE of a linear estimation of energy contour for V segment: Avg., Std., Skewness, Kurtosis
- 43-48 energy on the first voiced segment: Avg., Std., Max., Min., Skewness, Kurtosis
- 49-54 energy on the last voiced segment: Avg., Std., Max., Min., Skewness, Kurtosis
- 55-58 energy-contour for unvoiced segments: Avg., Std., Skewness, Kurtosis
- 59-62 Tilt of a linear estimation of energy contour for U segments: Avg., Std., Skewness, Kurtosis
- 63-66 MSE of a linear estimation of energy contour for U segments: Avg., Std., Skewness, Kurtosis
- 67-72 energy on the first unvoiced segment: Avg., Std., Max., Min., Skewness, Kurtosis
- 73-78 energy on the last unvoiced segment: Avg., Std., Max., Min., Skewness, Kurtosis
- 79 Voiced rate: Number of voiced segments per second
- 80-85 Duration of Voiced: Avg., Std., Max., Min., Skewness, Kurtosis
- 86-91 Duration of Unvoiced: Avg., Std., Max., Min., Skewness, Kurtosis
- 92-97 Duration of Pauses: Avg., Std., Max., Min., Skewness, Kurtosis
- 98-103 Duration ratios: Pause/(Voiced+Unvoiced), Pause/Unvoiced, Unvoiced/(Voiced+Unvoiced), Voiced/(Voiced+Unvoiced), Voiced/Pause, Unvoiced/Pause

Dynamic matrix is formed with 13 features computed for each voiced segment and contains

1-6. Coefficients of 5-degree Lagrange polynomial to model F0 contour

- 7-12. Coefficients of 5-degree Lagrange polynomial to model energy contour
- 13. Duration of the voiced segment

Dynamic prosody features are based on Najim Dehak, "Modeling Prosodic Features With Joint Factor Analysis for Speaker Verification", 2007

Script is called as follows

Examples command line:

```
>>> python prosody.py "../audios/" "prosodyfeaturesst.txt" "true" "false" "txt"
>>> python prosody.py "../audios/" "prosodyfeaturesst.csv" "true" "false" "csv"
>>> python prosody.py "../audios/" "prosodyfeaturesdyn.pt" "false" "false" "torch"
>>> python prosody.py "../audios/" "prosodyfeaturesdyn.csv" "false" "false" "csv"
```

Examples directly in Python

 $\textbf{extract_features_file} \ (\textit{audio}, \textit{static=True}, \textit{plots=False}, \textit{fmt='npy'}, \textit{kaldi_file=''})$

Extract the prosody features from an audio file

Parameters

- audio .wav audio file.
- **static** whether to compute and return statistic functionals over the feature matrix, or return the feature matrix computed over frames
- plots timeshift to extract the features

- **fmt** format to return the features (npy, dataframe, torch, kaldi)
- kaldi_file file to store kaldi features, only valid when fmt=="kaldi"

Returns features computed from the audio file.

extract_features_path (path_audio, static=True, plots=False, fmt='npy', kaldi_file='') Extract the prosody features for audios inside a path

Parameters

- path_audio directory with (.wav) audio files inside, sampled at 16 kHz
- **static** whether to compute and return statistic functionals over the feature matrix, or return the feature matrix computed over frames
- plots timeshift to extract the features
- **fmt** format to return the features (npy, dataframe, torch, kaldi)
- kaldi_file file to store kaldifeatures, only valid when fmt=="kaldi"

Returns features computed from the audio file.

plot_pros (data_audio, fs, F0, segmentsV, segmentsU, F0_features)
Plots of the prosody features

Parameters

- data_audio speech signal.
- **fs** sampling frequency
- **FO** contour of the fundamental frequency
- segmentsV list with the voiced segments
- segmentsU list with the unvoiced segments
- **FO_features** vector with f0-based features

Returns plots of the prosody features.

prosody_dynamic(audio)

Extract the dynamic prosody features from an audio file

Parameters audio – .wav audio file.

Returns array (N,13) with the prosody features extracted from an audio file. N= number of voiced segments

```
>>> prosody=Prosody()
>>> file_audio="../audios/001_ddk1_PCGITA.wav"
>>> features=prosody.prosody_dynamic(file_audio)
```

prosody_static (audio, plots)

Extract the static prosody features from an audio file

Parameters

- audio .wav audio file.
- plots timeshift to extract the features

Returns array with the 103 prosody features

```
>>> prosody=Prosody()
>>> file_audio="../audios/001_ddk1_PCGITA.wav"
>>> features=prosody.prosody_static(file_audio, plots=True)
```

CHAPTER

FIVE

PHONOLOGICAL FEATURES

Created on Jun 24 2020

@author: J. C. Vasquez-Correa

class phonological.Phonological

Compute phonological features from continuous speech files.

18 descriptors are computed, bases on 18 different phonological classes from the phonet toolkit https://phonet.readthedocs.io/en/latest/?badge=latest

It computes the phonological log-likelihood ratio features from phonet

Static or dynamic matrices can be computed:

Static matrix is formed with 108 features formed with (18 descriptors) x (6 functionals: mean, std, skewness, kurtosis, max, min)

Dynamic matrix is formed with the 18 descriptors computed for frames of 25 ms with a time-shift of 10 ms.

Script is called as follows

```
>>> python phonological.py <file_or_folder_audio> <file_features> <static (true_ 
or false)> <plots (true or false)> <format (csv, txt, npy, kaldi, torch)>
```

Examples command line:

```
>>> python phonological.py "../audios/" "phonologicalfeaturesst.txt" "true" "false

... " "txt"
>>> python phonological.py "../audios/" "phonologicalfeaturesst.csv" "true" "false

... "csv"
>>> python phonological.py "../audios/" "phonologicalfeaturesdyn.pt" "false"

... "false" "torch"
>>> python phonological.py "../audios/" "phonologicalfeaturesdyn.csv" "false"

... "false" "csv"
```

Examples directly in Python

extract_features_file (audio, static=True, plots=False, fmt='npy', kaldi_file='')

Extract the phonological features from an audio file

Parameters

- audio .wav audio file.
- **static** whether to compute and return statistic functionals over the feature matrix, or return the feature matrix computed over frames
- plots timeshift to extract the features
- **fmt** format to return the features (npy, dataframe, torch, kaldi)
- kaldi_file file to store kaldi features, only valid when fmt=="kaldi"

Returns features computed from the audio file.

extract_features_path (path_audio, static=True, plots=False, fmt='npy', kaldi_file='') Extract the phonological features for audios inside a path

Parameters

- path_audio directory with (.wav) audio files inside, sampled at 16 kHz
- **static** whether to compute and return statistic functionals over the feature matrix, or return the feature matrix computed over frames
- plots timeshift to extract the features
- **fmt** format to return the features (npy, dataframe, torch, kaldi)
- kaldi_file file to store kaldifeatures, only valid when fmt=="kaldi"

Returns features computed from the audio file.

CHAPTER SIX

NEED HELP?

If you have trouble with Disvoice, please write to Camilo Vasquez at: juan.vasquez@fau.de

REFERENCES

If you use Disvoice for research purposes, please cite the following papers, depending on the features you use:

7.1 glottal features

[1] Belalcázar-Bolaños, E. A., Orozco-Arroyave, J. R., Vargas-Bonilla, J. F., Haderlein, T., & Nöth, E. (2016, September). Glottal Flow Patterns Analyses for Parkinson's Disease Detection: Acoustic and Nonlinear Approaches. In International Conference on Text, Speech, and Dialogue (pp. 400-407). Springer.

7.2 phonation features

- [1] T. Arias-Vergara, J. C. Vásquez-Correa, J. R. Orozco-Arroyave, Parkinson's Disease and Aging: Analysis of Their Effect in Phonation and Articulation of Speech, Cognitive computation, (2017).
- [2] Vásquez-Correa, J. C., et al. "Towards an automatic evaluation of the dysarthria level of patients with Parkinson's disease." Journal of communication disorders 76 (2018): 21-36.

7.3 articulation features

- [1] Vásquez-Correa, J. C., et al. "Towards an automatic evaluation of the dysarthria level of patients with Parkinson's disease." Journal of communication disorders 76 (2018): 21-36.
- [2]. J. R. Orozco-Arroyave, J. C. Vásquez-Correa et al. "NeuroSpeech: An open-source software for Parkinson's speech analysis." Digital Signal Processing (2017).

7.4 prosody features

- [1]. N., Dehak, P. Dumouchel, and P. Kenny. "Modeling prosodic features with joint factor analysis for speaker verification." IEEE Transactions on Audio, Speech, and Language Processing 15.7 (2007): 2095-2103.
- [2] Vásquez-Correa, J. C., et al. "Towards an automatic evaluation of the dysarthria level of patients with Parkinson's disease." Journal of communication disorders 76 (2018): 21-36.

7.5 phonological features

[1] Vásquez-Correa, J. C., Klumpp, P., Orozco-Arroyave, J. R., & Nöth, E. (2019). Phonet: a Tool Based on Gated Recurrent Neural Networks to Extract Phonological Posteriors from Speech. Proc. Interspeech 2019, 549-553.

CHAPTER

EIGHT

INSTALLATION

From the source file:

git clone https://github.com/jcvasquezc/phonet
cd disvoice
bash install.sh

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CHAPTER

NINE

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HELP

If you have trouble with Disvoice, please write to Camilo Vasquez at: juan.vasquez@fau.de

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