TensorFlow

TensorFlow API r1.4

tf.contrib.ffmpeg.decode_audio

```
decode_audio(
    contents,
    file_format=None,
    samples_per_second=None,
    channel_count=None
)
```

Defined in tensorflow/contrib/ffmpeg_ops.py.

See the guide: FFmpeg (contrib) > Encoding and decoding audio using FFmpeg

Create an op that decodes the contents of an audio file.

Note that ffmpeg is free to select the "best" audio track from an mp4. https://trac.ffmpeg.org/wiki/Map

Args:

- contents: The binary contents of the audio file to decode. This is a scalar.
- file_format: A string or scalar string tensor specifying which format the contents will conform to. This can be mp3, mp4, ogg, or wav.
- samples_per_second: The number of samples per second that is assumed, as an **int** or scalar **int32** tensor. In some cases, resampling will occur to generate the correct sample rate.
- channel_count: The number of channels that should be created from the audio contents, as an int or scalar int32 tensor. If the contents have more than this number, then some channels will be merged or dropped. If contents has fewer than this, then additional channels will be created from the existing ones.

Returns:

A rank-2 tensor that has time along dimension 0 and channels along dimension 1. Dimension 0 will be samples_per_second * length_in_seconds wide, and dimension 1 will be channel_count wide. If ffmpeg fails to decode the audio then an empty tensor will be returned.

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