Compute part

The Idea of the project is to process the video recommendation tags for online YouTube videos.

Working of tag recommendation system

Steps

Step 1.

- 1. Pull a sample video from the YouTube API
- 2. Extract the audio from the Video
- 3. Convert audio to transcript (Text)
- 4. Use NLP libraries over the video transcript and extract the recommended tags for the videos
- 5. Compare the Tags with the title/ tags of the videos and show the analysis

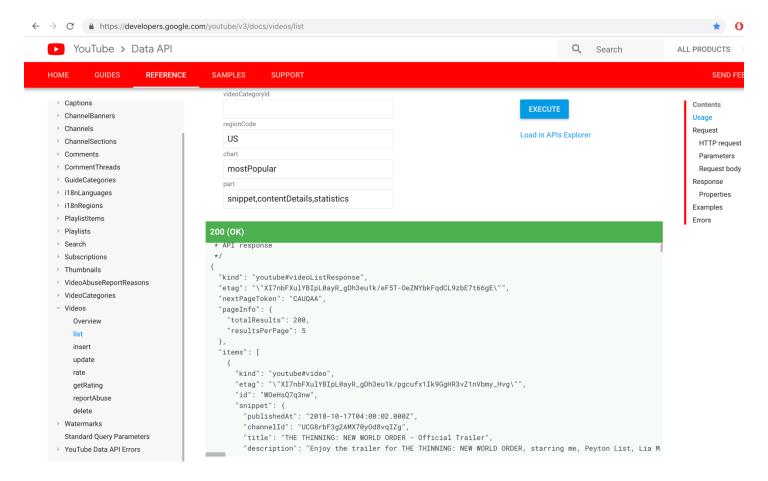
Analysis could be any of the information that is useful for the video owner.

Following will serve as the examples for analysis

- 1. Compare the video description and NLP recommended tags, show how far the title, video tags are deviated from the NLP recommended tags/ key words.
- 2. Looking at the deviation form step 1, system recommeds additional tags/ keywords which upon inclusion into the tags will improve the video when searched by user.

Example working demonstration of the Idea

Hit the YouTube API and collect a list of sample videos. For example I pulled most popular videos in US at this particular time



Step 2.

Pull a video ID from the API and fetch the video from youtube.

Extract the audio from the Video

Download the mp3 file from the YouTube API command as shown below

```
/m18-71:~$ youtube-dl --extract-audio --audio-format mp3 https://www.youtube.com/watch?v=W0eHsQ7q3nw
[youtube] WOeHsQ7q3nw: Downloading webpage
[youtube] WOeHsQ7q3nw: Downloading video info webpage
[download] Destination: THE THINNING - NEW WORLD ORDER - Official Trailer-WOeHsQ7q3nw.webm
[download] 100% of 2.35MiB in 00:00
[ffmpeg] Destination: THE THINNING - NEW WORLD ORDER - Official Trailer-WOeHsQ7q3nw.mp3
Deleting original file THE THINNING - NEW WORLD ORDER - Official Trailer-WOeHsQ7q3nw.webm (pass -k to keep)
skatta3@vm18-71:~$ ls
examples.desktop THE THINNING - NEW WORLD ORDER - Official Trailer-WOeHsQ7q3nw.mp3
```

Update the mp3 file to a intermediate convertible file to consume by the speech to text converter later

```
xamples.desktop speech.txt
                                               temp.log
ocketsphinx sphinxbase THE THINNING - NEW WORLD ORDER - Official Trailer-WOeHsQ7q3nw.mp3
skatta3@vm18-71:~$ ffmpeg -i THE\ THINNING\ -\ NEW\ WORLD\ ORDER\ -\ Official\ Trailer-WOeHsQ7q3nw.mp3 -ar 16000 -ac 1 file.wav
ffmpeg version 2.8.15-0ubuntu0.16.04.1 Copyright (c) 2000-2018 the FFmpeg developers
built with gcc 5.4.0 (Ubuntu 5.4.0-6ubuntu1~16.04.10) 20160609
configuration: --prefix=/usr --extra-version=0ubuntu0.16.04.1 --build-suffix=-ffmpeg --toolchain=hardened --libdir=/usr/lib/x86
configuration: --prefix=/usr --extra-version=@ubuntu0.10.04.1 --bulla-suffix=-ffmpeg --toolchain=haraenea --libair=/usr/lib/x86
nu --cc=cc --cxx=g++ --enable-gpl --enable-shared --disable-stripping --disable-decoder=libopenjpeg --disable-decoder=libschroed
le-gnutls --enable-ladspa --enable-libass --enable-libbluray --enable-libbs2b --enable-libcaca --enable-libcdio --enable-libflit
able-libfribidi --enable-libgme --enable-libgsm --enable-libmodplug --enable-libmp3lame --enable-libopenjpeg --enable-libopus --
edinger --enable-libshine --enable-libsnappy --enable-libsoxr --enable-libspeex --enable-libssh --enable-libtheora --enable-libt
--libmavpack --enable-libwebp --enable-libx265 --enable-libxvid --enable-libzvbi --enable-openal --enable-opengl --enable-x11grab
libzmq --enable-frei0r --enable-libx264 --enable-libopencv
                           54. 31.100 / 54. 31.100
  libavutil
                          56. 60.100 / 56. 60.100
56. 40.101 / 56. 40.101
56. 4.100 / 56. 4.100
  libavcodec
  libavformat
  libavdevice
  libavfilter
                            5. 40.101 /
                                                5. 40.101
                            2. 1. 0 / 2. 1. 0
3. 1.101 / 3. 1.101
1. 2.101 / 1. 2.101
  libavresample
  libswscale
  libswresample
                          53. 3.100 / 53. 3.100
  libpostproc
mp3 @ 0x1f75440] Skipping 0 bytes of junk at 237.
nput #0, mp3, from 'THE THINNING - NEW WORLD ORDER - Official Trailer-WOeHsQ7q3nw.mp3':
  Metadata:
     encoder
                               : Lavf56.40.101
  Duration: 00:02:42.22, start: 0.023021, bitrate: 103 kb/s
     Stream #0:0: Audio: mp3, 48000 Hz, stereo, s16p, 103 kb/s
     Metadata:
                                   : Lavc56.60
        encoder
Output #0, wav, to 'file.wav':
  Metadata:
     ISFT
                                : Lavf56.40.101
     Stream #0:0: Audio: pcm_s16le ([1][0][0] / 0x0001), 16000 Hz, mono, s16, 256 kb/s
     Metadata:
        encoder
                                   : Lavc56.60.100 pcm_s16le
tream mapping:
  Stream #0:0 -> #0:0 (mp3 (native) -> pcm_s16le (native))
 <u>ideo:0kB audio:5069kB su</u>btitle:0kB other streams:0kB global headers:0kB muxing overhead: 0.001503%
:katta3@vm18-71:~$ sudo pocketsphinx_continuous -infile file.wav > temp.log > speech.txt &
```

Step3.

As shown in above screenshot the last command converts the speech from the intermediate file to the text file with a log file in between.

We can see the output diff of the file while generated

```
skatta3@vm18-71:~$ cat speech.txt | tail -10
and
where
this
we are live in the memorial service that light reading all that out this governor dean reading los
landpoint high school
skatta3@vm18-71:~$ cat speech.txt | tail -10
and
where
this
we are live in the memorial service that light reading all that out this governor dean reading los
landpoint high school
smartphones the it known for you and try to ruin anything oversight committee for
i'm not going to be increased for haven't gotten around not working for him or not but the gun
the difference
```

The output file from here needs to be fed into the NLP module to note the diffs generated.

Data Intensive part

The above compute process is repeated for a huge number of videos and the results to be displayed consolidated in a dashboard.

The backend part of streaming the data is processed from Kafka engine and the data final text, NLP intermediate files are stored on Hadoop cluster