Text-Based Video Editing App — EasyUtube

4소 김동진 레쑤언휘 박세은 송민주 이은주 이종준

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Part 1.

Purpose & Necessity



1.1 Purpose & Necessity

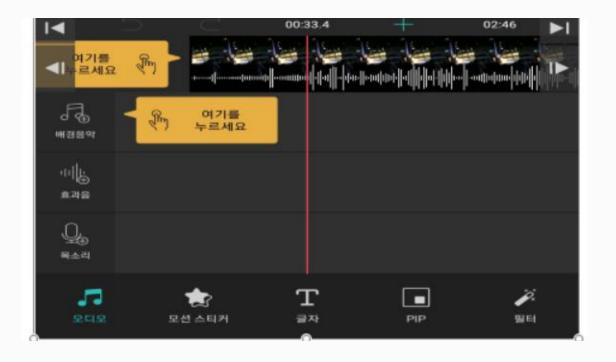
Disadvantages of the current video editing APP



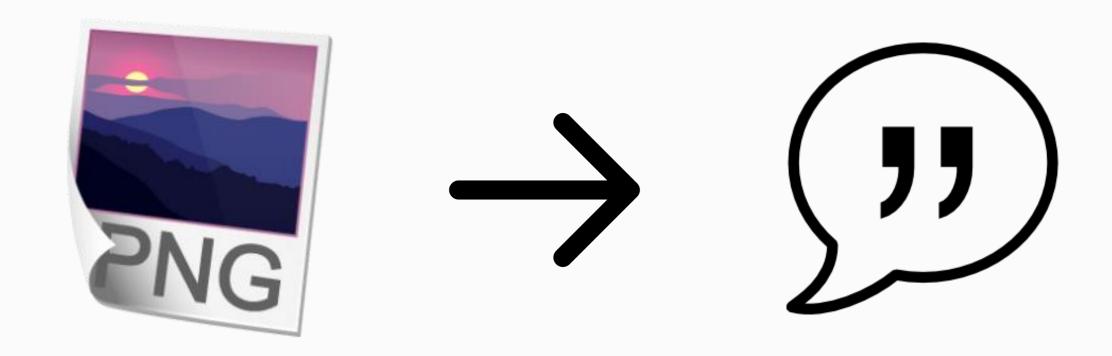
Ex) Viva Video(비바비디오)

1.1 Purpose & Necessity

Disadvantages of the current video editing APP



1.1 Purpose & Necessity



Converting voice signals in images to TEXT based rather than image based

Part 2.

Contents & Strategy



2.1 Contents

Android UI/UX

The UI/UX will be developed to fit the screen of the smartphone in the UI/UX of the existing clip/stick image

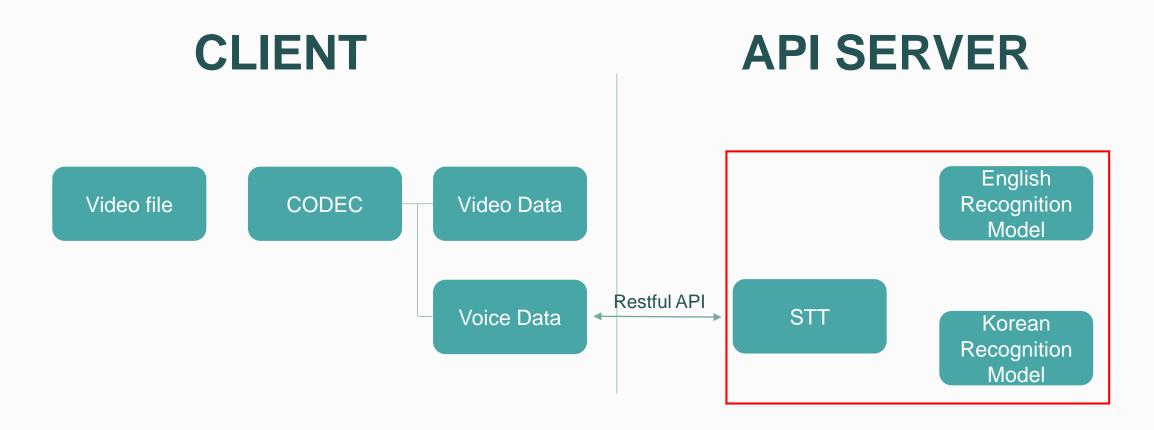
Video CODEC

It is planning to develop video CODEC that works on Android OS

Voice Recognition Device

Ability to convert English and Korean voice signals to TEXT and extract positional info

2.1 Contents

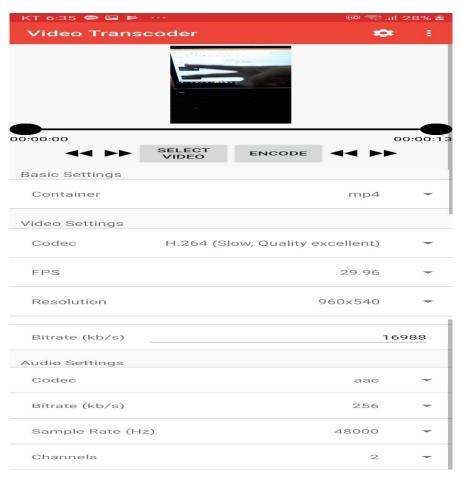


▲ System Configuration Diagram

(1) Video Application from Android apps

Play and Edit

- Use Open Source



(2) FFMPEG application from Android App

- Audio and video separation
- Audio and video decoding in progress
- Play and Edit

(2-1) FFMPEG

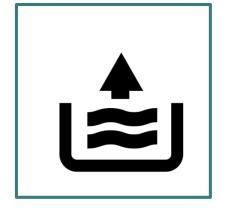
FFMPEG is a computer program that records and converts digital voice and video streams into various forms.

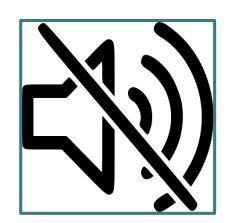


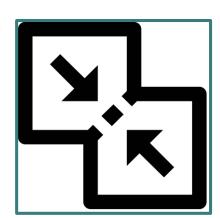
(2-2) Example of FFMPEG











(2-2) Example of FFMPEG

- 1) Import video for editing
- 2) Find the setting value of the cutting section

```
C:₩Windows₩System32₩cmd.exe
                 ₩AppData₩Roaming₩uTorrent>ffmpeg -i "Mnet 보이스 코리아.E07.12
3323.HDTV.H264.720p-MonG.avi" -f mp4 -vcodec mpeg4 -b 250000 -ar 24000 -ab 64 -a
 2 -ss 00:00:10.00 -t 3 avi2mp4example01.mp4
ffmpeg version N-38938-ge01f478 Copyright (c) 2000-2012 the Fmpeg developers
 built on Mar 19 2012 23:20:40 with gcc 4.6.2
 configuration: --enable-gpl --enable-version3 --disable-w32threads --enable-ru
ntime-cpudetect --enable-avisynth --enable-bzlib --enable-frei0r --enable-libope
ncore-amrnb --enable-libopencore-amrwb --enable-libfreetype --enable-libgsm --e
able-l<u>ibmp3l</u>ame --enable-libopenjpeg --enable-librtmp --enable-libschroedinger
enable-libspeex --enable-libtheora --enable-libvo-aacenc --enable-libvo-amrwben
 --enable-libvorbis --enable-libvpx --enable-libx264 --enable-libxavs --enable-
libxvid --enable-zlib
 libavutil
                51. 42.100 / 51. 42.100
 libavcodec
                54. 12.100 / 54. 12.100
 libavformat
                54. 2.100 / 54. 2.100
 libaudevice
                53. 4.100 / 53. 4.100
 libavfilter
                 2. 65.101 / 2. 65.101
 libswscale
                 2. 1.100 / 2. 1.100
 libswresample 0. 7.100 / 0. 7.100
                52. 0.100 / 52. 0.100
avi @ 000000000200F4E0] Switching to NI mode, due to poor interleaving
 avi @ 000000000200F4E0] max analyze duration 5000000 reached at 501600
```

(2-2) Example of FFMPEG

3) Cut the parts you want and create them to mp3 files

```
avi @ 000000000200F4E0] max_analuze_duration 5000000 reached at 5016000
Input #0, avi, from 'Mnet 蹂叫觉??肄禮一??E07.120323.HDTV.H264.720p-MonG.avi':
  Metadata:
                    : 장치
    title
                    : Lavf54.0.100
  Duration: 01:29:57.39, start: 0.000000, bitrate: 2642 kb/s
    Stream #0:0: Uideo: h264 (High) (H264 / 0x34363248), yuv420p, 1280x720 [SAR
1:1 DAR 16:9], 29.97 fps, 29.97 tbr, 29.97 tbn, 59.94 tbc
    Stream #0:1: Audio: mp3 (U[0][0][0] / 0x0055), 48000 Hz, stereo, s16, 128 kb
Please use -b:a or -b:v, -b is ambiguous
File 'avi2mp4example01.mp4' already exists. Overwrite ? [y/N] y
w:1280 h:720 pixfmt:yuv420p tb:1/1000000 sar:1/1 sws_param:
The bitrate parameter is set too low. It takes bits/s as argument, not kbits/s
Output #0, mp4, to 'avi2mp4example01.mp4':
  Metadata:
                    : 장치
    title
                    : Lauf54.2.100
    encoder
    Stream #0:0: Video: mpeg4 ( [0][0][0] / 0x0020), yuv420p, 1280x720 [SAR 1:1
DAR 16:9], q=2-31, 250 kb/s, 30k tbn, 29.97 tbc
    Stream #0:1: Audio: aac (@[0][0][0] / 0x0040), 24000 Hz, 2 channels, s16, 0
kb/s
Stream mapping:
 Stream #0:0 -> #0:0 (h264 -> mpeg4)
  Stream #0:1 -> #0:1 (mp3 -> libvo_aacenc)
Press [q] to stop, [?] for help
          0 fps=0.0 q=0.0 size=
                                      OkB time=00:00:00.00 bitrate= 0.0kbits/s
        13 fps= 13 q=21.2 size=
                                     139kB time=00:00:00.43 bitrate=2617.9kbits
```

(3) Text Conversion with Deep Learning

- Convert audio files to text using deep learning
- Use servers and clients
- Combine word-specific audio and video

(3) Deep Learning Techniques for Text Conversion

<Recuri

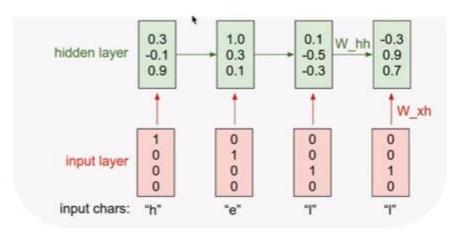
- model
- voice of nat

Character-level language model example

Vocabulary: [h,e,l,o]

Example training sequence: "hello"

$$h_t = anh(W_{hh}h_{t-1} + W_{xh}x_t)$$



e case

Fei-Fei Li & Andrej Karpathy & Justin Johnson

Lecture 10 - 20

8 Feb 2016

(4) Text-based video editing



Part 3.

Plan Of Processing



3.1 Plan

Period	Content
September Week 1-2	Team member composition and topic selection overall understanding and investigation of the subject Role sharing and implementation schedule plan by team member
September Week 3-5	Use ffmpeg in Android app
October Week 1-2	Play and edit Android app videos
October Week 3-5	intermediate inspection and fix bug Survey Deep Learning
November Week 1-3	Convert text using deep learning
November Week 4	Combine video by text
December Week 1	Modify Project Bug
December Week 2	Project testing and finishing Preparing Project Results Presentation

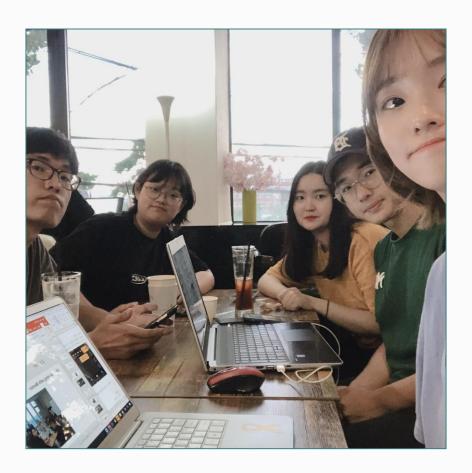
3.1 Plan

09/06 Meeting with Mentor



3.1 Plan

09/15 Conference



Part 4.

Expected Results



4.1 Expected Results







Thank You

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