# CV2 Examples

February 3, 2017

#### 1 Program 1 - horizortsfreq

Program to generate a horizontal spatial frequency, store it in a file, and display it on the screen \* Import numpy and cv2.

```
In []: import cv2
    import numpy as np
```

• 1000 sine values between 0 and 2pi\*f for f periods over the width of the picture, add a 1.0 to get all positive numbers, divide by 2 to get the normalized range between 0 and 1, which imwrite expects for floats:

• Diagonal matrix with sine wave on the diagonal:

```
In [ ]: d=np.diag(sinewave)
```

• Matrix with sinewave from left to right, identically on each row, with 500 rows:

```
In []: A=np.dot(np.ones((500,1000)),d)
```

• Display the resulting frame.

```
In []: cv2.imshow('Horizontale Ortsfrequenz', A)
```

• Write photo to jpg file. Mult with 255 because imwrite expects a range of 0...255:

```
In [ ]: cv2.imwrite('horizOrtsfreq.jpg', 255*A)
```

• Keep photo window open until key 'q' is pressed:

• When everything done, release the capture

```
In [ ]: cv2.destroyAllWindows()
```

#### 2 Program 2 - pyrecfftanimation

Using Pyaudio, record sound from the audio device and plot the fft magnitude spectrum life, for 8 seconds. Usage example:

python pyrecfftanimation.py

Import relevant libraries

• Defining the variables.

```
In []: CHUNK = 2048 #Blocksize
    WIDTH = 2 #2 bytes per sample
    CHANNELS = 1 #2
    RATE = 32000 #Sampling Rate in Hz
    RECORD_SECONDS = 70

fftlen=CHUNK/2
```

• Set up for the plot.

• Function to process the audio blockwise to plot live spectrum of it.

```
In []: def animate(i):
    # update the data
    #Reading from audio input stream into data with block length "CHUNK":
    data = stream.read(CHUNK)
    #Convert from stream of bytes to a list of short integers (2 bytes here) in "samples
    #shorts = (struct.unpack( "128h", data ))
    shorts = (struct.unpack( 'h' * CHUNK, data ));
    samples=np.array(list(shorts),dtype=float);

#plt.plot(samples) #<-- here goes the signal processing.
line.set_ydata(20.0*np.log((np.abs(np.fft.fft(samples[0:fftlen])/np.sqrt(fftlen))+1)
    #line.set_ydata(samples)
    return line,</pre>
```

Initialising the data to be represented.

Initialize the soundcard(audio port) for input and output operations.

```
In [ ]: p = pyaudio.PyAudio()
```

• Print out device information about input channels and their corresponding sampling rate.

• Define the stream of data to be processed for input and output on audio port.

• Real time plot.

• When everything done, release the capture.

```
In []: print("* done")

    f.close()
    stream.stop_stream()
    stream.close()
```

### 3 Program 3 - pyrecspecwaterfall

Using Pyaudio, record sound from the audio device and plot a waterfall spectrum display, for 8 seconds. Usage example:

python pyrecspecwaterfall.py

• Import the relevant modules.

```
In []: import pyaudio
    import struct
    import numpy as np
    import cv2
```

• Define the variables.

```
In []: CHUNK = 1024 #Blocksize
    WIDTH = 2 #2 bytes per sample
    CHANNELS = 1 #2
    RATE = 32000 #Sampling Rate in Hz
```

• Initialise the sound card(audio port) and print out the device information.

```
In []: p = pyaudio.PyAudio()

a = p.get_device_count()
print("device count=",a)

for i in range(0, a):
    print("i = ",i)
    b = p.get_device_info_by_index(i)['maxInputChannels']
    print(b)
    b = p.get_device_info_by_index(i)['defaultSampleRate']
    print(b)
```

• Define the stream and its parameters.

• Start recording and plot the real time waterfall(in opposite direction), where the horizontal axis is the magnitude and vertical axis is the time.

```
In [ ]: while(True):
```

• Reading from audio input stream into data with block length "CHUNK":

```
In [ ]: data = stream.read(CHUNK)
```

• Convert from stream of bytes to a list of short integers (2 bytes here) in "samples": In []: shorts = (struct.unpack( 'h' \* CHUNK, data ));

```
samples=np.array(list(shorts),dtype=float);
```

• Shift "frame" 1 up:

```
In []:
           frame[0:(rows-1),:]=frame[1:rows,:];
```

• Compute magnitude of 1D FFT of sound with suitable normalization for the display:write magnitude spectrum in lowes row of "frame":

```
R=0.25*np.log((np.abs(np.fft.fft(samples[0:fftlen])[0:(fftlen/2)]/np.sqrt(fftlen))+10.25*np.log((np.abs(np.fft.fft(samples[0:fftlen])[0:(fftlen/2)]/np.sqrt(fftlen))+10.25*np.log((np.abs(np.fft.fft(samples[0:fftlen])[0:(fftlen/2)]/np.sqrt(fftlen))+10.25*np.log((np.abs(np.fft.fft(samples[0:fftlen])[0:(fftlen/2)]/np.sqrt(fftlen))+10.25*np.log((np.abs(np.fft.fft(samples[0:fftlen])[0:(fftlen/2)]/np.sqrt(fftlen))+10.25*np.log((np.abs(np.fft.fft(samples[0:fftlen])[0:(fftlen/2)]/np.sqrt(fftlen))+10.25*np.log((np.abs(np.fft.fft(samples[0:fftlen])[0:(fftlen/2)]/np.sqrt(fftlen))+10.25*np.sqrt(fftlen))+10.25*np.sqrt(fftlen))+10.25*np.sqrt(fftlen))+10.25*np.sqrt(fftlen))+10.25*np.sqrt(fftlen))+10.25*np.sqrt(fftlen))+10.25*np.sqrt(fftlen))+10.25*np.sqrt(fftlen))+10.25*np.sqrt(fftlen))+10.25*np.sqrt(fftlen))+10.25*np.sqrt(fftlen))+10.25*np.sqrt(fftlen))+10.25*np.sqrt(fftlen))+10.25*np.sqrt(fftlen))+10.25*np.sqrt(fftlen))+10.25*np.sqrt(fftlen))+10.25*np.sqrt(fftlen))+10.25*np.sqrt(fftlen))+10.25*np.sqrt(fftlen))+10.25*np.sqrt(fftlen))+10.25*np.sqrt(fftlen))+10.25*np.sqrt(fftlen))+10.25*np.sqrt(fftlen))+10.25*np.sqrt(fftlen))+10.25*np.sqrt(fftlen))+10.25*np.sqrt(fftlen))+10.25*np.sqrt(fftlen))+10.25*np.sqrt(fftlen))+10.25*np.sqrt(fftlen))+10.25*np.sqrt(fftlen))+10.25*np.sqrt(fftlen))+10.25*np.sqrt(fftlen))+10.25*np.sqrt(fftlen))+10.25*np.sqrt(fftlen))+10.25*np.sqrt(fftlen))+10.25*np.sqrt(fftlen))+10.25*np.sqrt(fftlen))+10.25*np.sqrt(fftlen))+10.25*np.sqrt(fftlen))+10.25*np.sqrt(fftlen))+10.25*np.sqrt(fftlen))+10.25*np.sqrt(fftlen))+10.25*np.sqrt(fftlen))+10.25*np.sqrt(fftlen))+10.25*np.sqrt(fftlen))+10.25*np.sqrt(fftlen))+10.25*np.sqrt(fftlen))+10.25*np.sqrt(fftlen))+10.25*np.sqrt(fftlen))+10.25*np.sqrt(fftlen))+10.25*np.sqrt(fftlen))+10.25*np.sqrt(fftlen))+10.25*np.sqrt(fftlen))+10.25*np.sqrt(fftlen))+10.25*np.sqrt(fftlen))+10.25*np.sqrt(fftlen))+10.25*np.sqrt(fftlen))+10.25*np.sqrt(fftlen))+10.25*np.sqrt(fftlen))+10.25*np.sqrt(fftlen))+10.25*np.sqrt(fftlen))+10.25*np.sqrt(fftlen))+10.25*np.sqrt(fftlen))+10.25*np.sqrt(fftlen
In []:
```

• Color mapping: Red:

```
In []:
            frame [rows-1,:,2]=R
```

• Green:

```
In []:
            frame [rows-1,:,1] = np.abs(1-2*R)
```

• Blue:

```
In []:
            frame [rows-1,:,0]=1.0-R
```

• Display the resulting frame

```
In []:
           cv2.imshow('frame',frame)
```

• Keep window open until key 'q' is pressed:

```
In []:
            if cv2.waitKey(1) & OxFF == ord('q'):
                break
```

• When everything done, release the capture

```
In [ ]: cv2.destroyAllWindows()
        stream.stop_stream()
        stream.close()
        p.terminate()
```

#### 4 Program 4 - videofft0ifftresampley

Program to capture a video from a camera, compute the Y-component, downsample it by a factor of N horizontally and vertically, and display it live on the screen. \* Import numpy and cv2.

```
In []: import numpy as np
    import cv2

cap = cv2.VideoCapture(0)
```

• Downsampling factor N:

 Mask to set to zero the 7/8 highest frequencies, only kep the 1/8 lowest frequencies in each direction: For rows:

• For columns:

Together

```
In [ ]: M=np.dot(Mr,Mc)
In [ ]: while(True):
```

• Encoding side: Capture frame-by-frame

```
In []: [ret, frame] = cap.read()
```

• Berechnung der Luminanz-Komponente Y: Y= 0.114*B*+0.587G+0.299\*R : /256 because the result is float values which imshow expects in range 0...1:

```
In []: Y=(0.114*frame[:,:,0]+0.587*frame[:,:,1]+0.299*frame[:,:,2])/256
```

• Downgesamplets Y, nur jedes Nte pixel horizontal und vertikal wir uebertragen:

```
In [ ]: #2D-FFT of Y
X=np.fft.fft2(Y)
```

• Set to zero the 7/8 highest spacial frequencies in each direction:

```
In [ ]:
            X = X * M
   • inverse 2D-FFT:
In [ ]:
            Y0=np.abs(np.fft.ifft2(X))
   • Downgesamplets Y0, nur jedes Nte pixel horizontal und vertikal wir uebertragen:
In []: DsO[0::N,::N] = YO[0::N,::N]
   • Decoding Side: Lowpass filter the sampled frame:
In [ ]: DsOfft=np.fft.fft2(DsO)
   • Set to zero the 7/8 highest spacial frequencies in each direction:
In [ ]:
            Ds1fft=Ds0fft*M
  • Inverse 2D-FFT:
In []:
            Yrek=np.abs(np.fft.ifft2(Ds1fft))
   • Display the resulting frames
In []:
            cv2.imshow('Decoder: Gefilterte Sampled Frames', Yrek*N*N)
            cv2.imshow('Decoder: FFT Bereich nach Null Setzen',np.abs(Ds1fft)/100)
            cv2.imshow('Decoder: FFT Bereich der Sampled Frames ',np.abs(DsOfft)/100)
            cv2.imshow('Encoder: Sampled Frames',Ds0)
            cv2.imshow('Encoder: Video nach Nullsetzen der hohen Ortsfrequenzen', YO)
            cv2.imshow('Encoder 2D FFT von Y und Null Setzen',np.abs(X)/(480))
            #cv2.imshow('Original', frame)
            cv2.imshow('Encoder: Luminanz Y',Y)
   • Press 'q' to close all the windows.
In []:
            if cv2.waitKey(1) & OxFF == ord('q'):
                 break
   • When everything done, release the capture
In [ ]: cap.release()
        cv2.destroyAllWindows()
```

#### 5 Program 5 - videofft0ifftresampleykey

Program to capture a video from a camera, compute the Y-component, downsample it by a factor of N horizontally and vertically, and display it live on the screen With keyboard switchable low pass filter and samplingkey f toggles the low pass filter, key s the samplingWith explanation text and state display in the image windows.

• Import numpy and cv2.

• Downsampling factor N:

• Mask to set to zero the 7/8 highest frequencies, only kep the 1/8 lowest frequencies in each direction:For rows:

```
In []: Mr=np.ones((r,1))

Mr[(r/8.0):(r-r/8.0),0]=np.zeros((3.0/4.0*r))
```

• For columns:

In [ ]: while(True):

```
In []: Mc=np.ones((1,c))

Mc[0,(c/8.0):(c-c/8)]=np.zeros((3.0/4.0*c))
```

```
• Together:
In []: M=np.dot(Mr,Mc)

ytext=np.zeros((rows,cols))
cv2.putText(ytext,"Down- and upsampling and LP filtering Demo", (20,50), cv2.FONT_HERSHEY
cv2.putText(ytext,"Toggle LP filter in 2D-FFT on/off: key f", (20,100), cv2.FONT_HERSHEY
cv2.putText(ytext,"Toggle sampling on/off: key s", (20,150), cv2.FONT_HERSHEY_SIMPLEX, 0
cv2.putText(ytext,"Quit: key q", (20,200), cv2.FONT_HERSHEY_SIMPLEX, 0.8, (255))

filteron=False
samplingon=False
```

• Encoding side: Capture frame-by-frame

• Berechnung der Luminanz-Komponente Y: Y= 0.114*B*+0.587G+0.299\*R : /256 because the result is float values which imshow expects in range 0...1:

• 2D-FFT of Y

```
In [ ]: X=np.fft.fft2(Y)
```

• Set to zero the 7/8 highest spacial frequencies in each direction:

```
In \lceil \rceil: X=X*M
```

• inverse 2D-FFT:

```
In []: Y=np.abs(np.fft.ifft2(X))

if samplingon==True:
```

• Downgesamplets Y0, nur jedes Nte pixel horizontal und vertikal wir uebertragen:

```
In [ ]: Y0=np.zeros((rows,cols));
Y0[0::N,::N]=Y[0::N,::N];
Y=Y0.copy()
```

#print("filter off")

• Decoding Side Make text:

```
In []: ytext2=np.zeros((rows,cols))
    if samplingon:
        cv2.putText(ytext2, "Sampling on", (20,20), cv2.FONT_HERSHEY_SIMPLEX, 0.8, (0.9))
        #print("sampling on")
else:
        cv2.putText(ytext2, "Sampling off", (20,20), cv2.FONT_HERSHEY_SIMPLEX, 0.8, (0.9))
        #print("sampling off")

if filteron:
        cv2.putText(ytext2, "Filter on", (20,50), cv2.FONT_HERSHEY_SIMPLEX, 0.8, (255))
        #print("filter on")
else:
        cv2.putText(ytext2, "Filter off", (20,50), cv2.FONT_HERSHEY_SIMPLEX, 0.8, (255))
```

• 2D-DFT:

```
In []:
            Dsfft=np.fft.fft2(Y)
            if filteron == True:
                #Lowpass filter the sampled frame:
                \#Dsfilt=N*scipy.siqnal.convolve2d(Ds0,filt,mode='same')
                #Set to zero the 7/8 highest spacial frequencies in each direction:
               Dsfft=Dsfft*M
            #scaling to maintain the energy after sampling and filtering
            if samplingon and filteron:
                Dsfft=Dsfft*N*N
            cv2.imshow('2D Discrete Fourier Transform of (downsampled, filtered) Luminance Y',np
   • Inverse 2D-FFT
In []:
            Y=np.abs(np.fft.ifft2(Dsfft))
            cv2.imshow('Decoder: reconstructed Luminance Y', Y+ytext2)

    Key Inputs: - 's' to toggle sampling. - 'f' to toggle filtering. - Press the key "q" to quit window.

In [ ]:
            key=cv2.waitKey(1) & OxFF;
            if key == ord('s'):
               samplingon = not samplingon;
            if key == ord('f'):
               filteron = not filteron;
            if key == ord('q'):
                break
   • When everything done, release the capture
In [ ]: cap.release()
        cv2.destroyAllWindows()
```

## 6 Program 6 - videofiltdisp

Program to capture a video from a camera, filter is, and display it live on the screen \* Import the following modules cv2, numpy and scipy.signal.

```
In []: import numpy as np
    import scipy.signal
    import cv2

cap = cv2.VideoCapture(0)
    [retval, frame] = cap.read()
    [r,c,d]=frame.shape
    print(r,c)
```

• 2D-Mask to set to zero the 7/8 highest frequencies, only kep the 1/8 lowest frequencies in each direction:For rows:

```
In []: Mr=np.ones((r,1))
        Mr[(r/8.0):(r-r/8.0),0]=np.zeros((3.0/4.0*r))
  • For columns
In [ ]: Mc=np.ones((1,c))
        Mc[0,(c/8.0):(c-c/8)]=np.zeros((3.0/4.0*c))
   • Together:
In []: M=np.dot(Mr,Mc)
   • Conmpute space-domain/inverse 2D Fourier transform of Low Pass filter:
In [ ]: h=np.abs(np.fft.ifft2(M))
        hc=np.concatenate((h[:,(c/2):c],h[:,0:(c/2)]),axis=1)
        hc=np.concatenate((hc[(r/2):r,:],hc[0:(r/2),:]))
   • Only keep the piece with the biggest components:
In []: hc=hc[(r/2-10):(r/2+10),(c/2-10):(c/2+10)]
        filt=hc
        while(True):
   • Capture frame-by-frame
In []:
            [ret, frame] = cap.read()
            Y=(0.114*frame[:,:,0]+0.587*frame[:,:,1]+0.299*frame[:,:,2])/256;
            Yfilt=scipy.signal.convolve2d(Y,filt,mode='same')
   • Display the resulting filtered frame
            cv2.imshow('Y low-pass filtered', Yfilt)
In [ ]:
            if cv2.waitKey(1) & OxFF == ord('q'):
                break
   • When everything done, release the capture
```

In [ ]: cap.release()

cv2.destroyAllWindows()

#### 7 Program 7 - videorecfft0ifftdisp

Program to capture a video from the default camera (0), compute the 2D FFT on the Green component, take the magnitude (phase) and display it live on the screen \* Import cv2 and numpy.

• Get size of frame:

 Mask to set to zero the 3/4 highest frequencies, only kep the 1/4 lowest frequencies in each direction: For rows:

• For columns:

```
In []: Mc=np.ones((1,c))

Mc[0,(c/8.0):(c-c/8)]=np.zeros((3.0/4.0*c))
```

• Together:

```
In [ ]: M=np.dot(Mr,Mc)
In [ ]: while(True):
```

• Capture frame-by-frame

• Compute magnitude of 2D FFT of green component with suitable normalization for the display:

```
In []: X=np.fft.fft2(frame[:,:,1]/255.0)
```

• Set to zero the 7/8 highest spacial frequencies in each direction:

• Display the resulting frame

```
In []: cv2.imshow('2D-FFT mit Null Setzen der hoechsten Ortsfrequenzen',frame)
```

• Inverse FFT, with abs to turn complex into float numbers:

• Keep window open until key 'q' is pressed:

• When everything done, release the capture

#### 8 Program 8 - videoresampley

Program to capture a video from a camera, compute the Y-component, downsample it by a factor of N horizontally and vertically, and display it live on the screen. \* Import numpy and cv2.

• Downsampling factor N:

In [ ]: while(True):

Capture frame-by-frame

```
In []: [ret, frame] = cap.read()
```

• Berechnung der Luminanz-Komponente Y: Y= 0.114*B*+0.587G+0.299\*R : /256 because the result is float values which imshow expects in range 0...1:

```
In []: Y=(0.114*frame[:,:,0]+0.587*frame[:,:,1]+0.299*frame[:,:,2])/256;
```

• Downgesamplets Y, nur jedes Nte pixel horizontal und vertikal wir uebertragen:

```
In []: Ds[0::N,::N]=Y[0::N,::N];
```

• Display the resulting frame

• Press 'q' to quit the windows.

```
In []: if cv2.waitKey(1) \& OxFF == ord('q'): break
```

• When everything done, release the capture