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```
%RTL_FM_PLL
%
% Simple FM radio receiver using phase-locked loop
% No stereo, no de-emphasis filter
%
% By R.W.

clear all, close all

% Script to set some essential parameters you need to figure yourself:
% - low-pass filter FLOW and decimator factor NDEC
% - Loop filter and its coefficients
% - Phase detector
% - voltage controlled oscillator
%rw_fmrx_init
```

Radio parameters

FM transmitter

```
%expFreq = 89.5e6;
% YLE 1
%expFreq = 87.9e6;
% YLE Puhe
expFreq = 103.7e6;
% YLE Radio Suomi
%expFreq = 94e6;

% Front-end sampling rate etc. Change the numbers at will
FESR = 240e3;
nSample = 4092*8;
nFrame = 12e2/8;

NDEC = 6;
fmax = 40e+03;
nyq = fmax/2;
```

Task & Explanation: Describe briefly the algorithm used in the filter design function The algorithm used is 48th order with frequency magnitude characteristics specified by how we set out cutoff frequency, in the second vector containing the desired magnitude response at each of the points specified in the vector where cutoff frequency is specified. Frequency sampling-based FIR filter design is adopted where FLOW holds the values for filter coefficients which consequently is used to filter The FM receiver code is made in

such a way, the signal processing blocks are firstly, it performs Phase Locked Loop, followed by filtering and finally decimating it.

```
FLOW = fir2(48, [0 15e3/nyq 17e3/nyq 1], [1 1 0 0]);

hSDRrRx = comm.SDRRTLReceiver(...
    'RadioAddress', '0',...
    'CenterFrequency', expFreq, ...
    'EnableTunerAGC', true, ...
    'SampleRate', FESR, ...
    'SamplesPerFrame', nSample, ...
    'FrequencyCorrection', 70, ...
    'OutputDataType', 'double')
fprintf('\n')

hSpectrumAnalyzer = dsp.SpectrumAnalyzer(...
    'Name', 'Received signal',...
    'Title', 'Received signal', ...
    'SpectrumType', 'Power density',...
    'FrequencySpan', 'Full', ...
    'SampleRate', FESR, ...
    'YLimits', [-60,0],...
    'SpectralAverages', 10, ...
    'FrequencySpan', 'Start and stop frequencies', ...
    'StartFrequency', -50e3, ...
    'StopFrequency', 50e3,...
    'Position', figposition([50 30 30 40]));

% Check out low-pass filtered signal if necessary
hSA3 = clone(hSpectrumAnalyzer);
set(hSA3, 'Name', 'Filtered signal', 'Title', 'filtered signal');

hAudio = audioDeviceWriter(FESR/NDEC, 'BufferSize', ceil(nSample*2/
NDEC));
getAudioDevices(hAudio);

% Received FM signal
rxfm = zeros(nSample,1);
% Initialization of VCO and, previous phase difference, filter memory
vcoph = 0; dphprev=0; lstate=0;

hSDRrRx =

comm.SDRRTLReceiver with properties:

    RadioAddress: '0'
CenterFrequency: 103700000
  EnableTunerAGC: true
      SampleRate: 240000
    OutputDataType: 'double'
    SamplesPerFrame: 32736
FrequencyCorrection: 70
  EnableBurstMode: false
```

Stream Processing

```
if ~isempty(sdrinfo(hSDRRx.RadioAddress))

    fprintf('Receive time %f [s]   \n',nSample/FESR*nFrame)

    memo = zeros(1, length(FLOW)-1);
    dphprev=0;
    for iFrame = 1 : nFrame
        rxSig = step(hSDRRx);
        rxSig = rxSig - mean(rxSig); % Remove DC component

        % Display received frequency spectrum
        hSpectrumAnalyzer(rxSig);

        % Optionally, low-pass filter before the PLL operation
        % Here, low-pass filtering is done after PLL.
        [rxfilt,memo]=filter(FLOW,1,rxSig,memo);
        %hSA3(rxfilt);

        % The loop operates at front-end sampling rate
        for ii = 1:nSample
            % Phase detector
            dph = rw_phdetector(rxSig(ii), vcoph);

            % First-order IIR filter as in the slides
            %rxfm(ii) = rw_loopf(ALPHA,dph,dphprev, lstate);
            rxfm(ii) = rw_loopf(dph,dphprev, lstate);
            dphprev = dph;
            lstate = rxfm(ii);

            % "NCO - numerically controlled oscillator"
            vcoph = rw_integrate(rxfm(ii), vcoph);
        end

        % Reduce noise level and downsample to Audio
        % Loop filter is low-pass but its stop-band attenuation is poor
        % Filtering done after PLL, attempt was made to filter it
        before
        % PLL,
        [rdfilt,memo]=filter(FLOW,1,rxfm,memo);
        rdec = rdfilt(1:NDEC:end);
        % AGC (automatic gain control). Seems it's not needed
        %rdecnorm = rdec/max(abs(rdec));

        % Underrun may occur in the loop
        nUnderrun = hAudio(rdec);
        if nUnderrun > 0
            fprintf('Audio player queue underrun by %d samples.
\n',nUnderrun);
```

```

        end
    end
else
    warning(message('SDR:sysobjdemos:MainLoop'))
end

Receive time 20.460000 [s]

```

Release all System objects

```

release(hSDRrRx);
clear hSDRrRx
release(hAudio)

%performing functions
function dph = rw_phdetector(rxsig,vcoph)
%%Strangely, an observation made here is that I was not able to receipt
%%anything when I used the model in Slide 14 of PLL lecture, later in
%%co-ordination with Gabriel(other team), I implemented model in Slide
15
%dph = real(rxsig) .* (-sin(vcoph)); According to Slide 14
    dph = imag(rxsig*exp(-1i*vcoph));

end
% First-order IIR filter as in the slides
function rxfm = rw_loopf(dph,dphprev, lstate)

    kp = 0.7 ;
    ki = 0.75;
    rxfm = kp*dph+ki*dph-kp*dphprev+ lstate;
end

function vcoph = rw_integrate(rxfm, vcoph)
    vcoph = rxfm + vcoph;
end

Audio player queue underrun by 207328 samples.
Audio player queue underrun by 201872 samples.
Audio player queue underrun by 185504 samples.
Audio player queue underrun by 190960 samples.
Audio player queue underrun by 185504 samples.
Audio player queue underrun by 185504 samples.
Audio player queue underrun by 185504 samples.
Audio player queue underrun by 201872 samples.
Audio player queue underrun by 190960 samples.
Audio player queue underrun by 212784 samples.
Audio player queue underrun by 212784 samples.
Audio player queue underrun by 180048 samples.
Audio player queue underrun by 289168 samples.
Audio player queue underrun by 234608 samples.
Audio player queue underrun by 201872 samples.
Audio player queue underrun by 185504 samples.

```

[illegible]

Audio player queue underrun by 174592 samples.
Audio player queue underrun by 201872 samples.
Audio player queue underrun by 180048 samples.
Audio player queue underrun by 174592 samples.
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Audio player queue underrun by 185504 samples.
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Audio player queue underrun by 180048 samples.
Audio player queue underrun by 250976 samples.
Audio player queue underrun by 343728 samples.
Audio player queue underrun by 267344 samples.
Audio player queue underrun by 229152 samples.
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Audio player queue underrun by 256432 samples.
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Published with MATLAB® R2018a