Software Defined Radio in Max/MSP

Tutorial 1 – System Design, IQ Signals, Mixing

This set of four tutorials explores the process of building a software defined radio receiver in Max/MSP.

The patches work with any baseband IQ signal source, including hardware devices and IQ data files.

The tutorials assume a basic understanding of software defined radio, receiver design, and experience with Max/MSP.

Mac OS drivers are provided for the Soft66lc2 and Funcube devices.

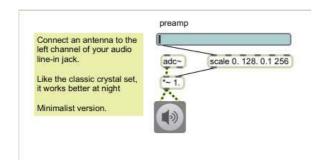
Important instructions

- Add the folder containing the tutorial files to the Max user file path in: Options | File Preferences.
- If possible set your audio sample rate to 96 KHz for the tutorial patches (in: Options | DSP Status) It improves bandwidth and audio quality.

The smallest radio

Signals are everywhere. In the 1965 TV series "Gilligans Island" an accidental blow to the head caused Gilligan's tooth fillings to act as a radio receiver.

Open the patch: minimal.maxpat



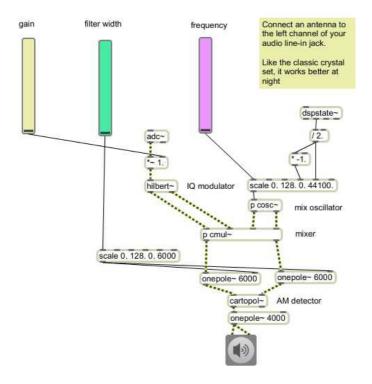
This patch passes audio from input to output with some amplification. If you hook a microphone to the input, you'll have a PA system. If you connect a dipole antenna to the

input you'll have a radio. I have listened to Radio Havana with this patch. There's no control over frequency or selectivity.

If you work with audio and live in an urban area you've probably had the experience of unwanted radio signals getting into your audio gear. The wiring becomes an inadvertent antenna. That's really all that's going on here.

If you have an antenna, connect it to your computer's audio input jack. Select line-in as the audio source. It works best with shielded transmission (coaxial cable). Also try it on a winter night under a full moon when signal propagation is optimum for low frequencies.

You may also want to try the patch: crystalSet.maxpat, which simulates a classic crystal set radio along the same lines as the previous patch.

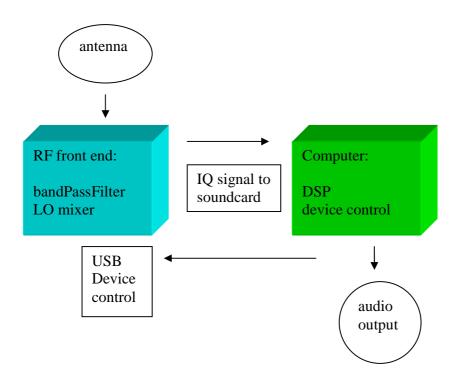


Note: please close any open patches before moving on to the next section.

System Design

In a software defined radio system (SDR), components traditionally built with hardware, are implemented in software using digital signal processing (DSP). Familiar products like cell phones and wireless routers are examples of SDR's. The project described here is a software defined radio receiver with three physical components:

- Antenna
- RF front-end
- computer



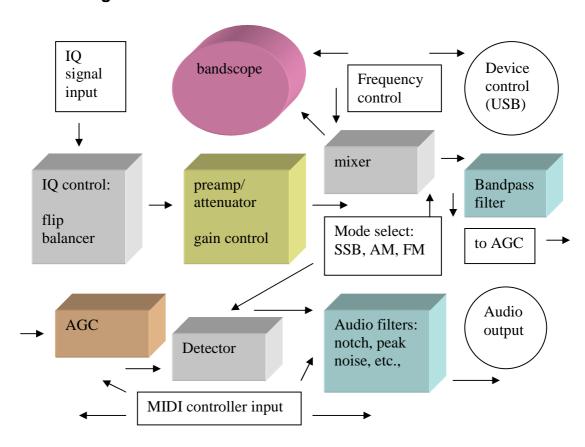
In a direct conversion SDR, the RF front end sends input signals through a bandpass filter to a mixer which shifts the frequency down into the audio spectrum. The mixer uses two local oscillator (LO) signals that are 90 degrees out of phase to produce "I" and "Q" signals (in-phase and quadrature). IQ signals are also referred to as complex or analytic signals. Using IQ signals simplifies frequency shifting and demodulation.

In these tutorials we'll use baseband IQ data files pre-recorded from SDR devices.

The computer reads baseband IQ signals, via the sound card, and converts them to digital data (ADC). The rest of the processing happens in software. This includes mixing, filtering and demodulation. The final result is converted to an analog audio by the sound card (DAC).

Typically an SDR program also sets device frequency and selects appropriate hardware filters via a serial (USB) interface.

Software Signal Path

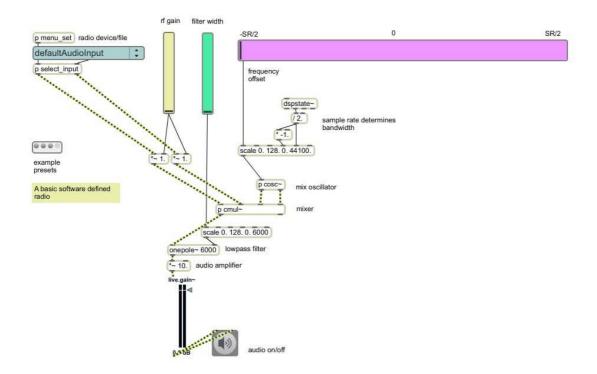


A Basic Radio

Sample IQ data files have been provided for this tutorial. Looping these files simulates signal input from a hardware device.

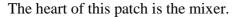
Note: If you have a hardware device capable of generating baseband IQ signals - and the software to control its frequency - it should work with the tutorial patches. Just connect it to the computer's audio input.

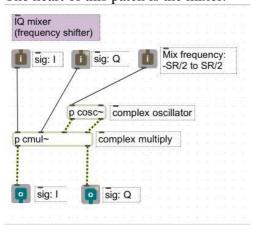
Open the patch: basicSDR1.maxpat



Start the audio by clicking the speaker icon [ezdac~] at the bottom of the patch. Then select one of the example preset buttons. (They're the miniature dots inside a gray rectangle). The example data (10meter96.wav) is from a 28 MHz amateur radio SSB contest. Try adjusting the frequency slider and filter width slider.

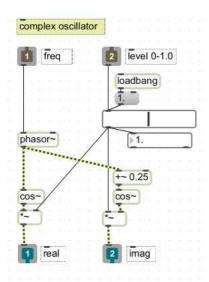
Let's follow the signal path starting in the upper left corner. The [umenu] selects an input source, in this case a looped IQ data file. Next the RF gain slider amplifies or attenuates incoming signals to a reasonable level.





The mixer shifts the center frequency of a baseband signal by complex multiplying it with output of a mix oscillator. The mix oscillator is a variable complex oscillator

[cosc~], controlled by the frequency slider. It produces a frequency ranging from –SR/2 to SR/2 (where SR is the current sample rate).



The mixer output frequency is the sum of the mix oscillator and the baseband signal. So essentially, the mixer acts like a tuning dial (within the passband). Sample rate determines passband width. With IQ signals sample rate is passband width. At this point, if you're new to the world of radio, I hope you have questions about negative frequencies and the apparent disregard for Nyquist.

A lowpass filter [onepole~] is next in the signal path. An SSB signal has a bandwidth less than 3 KHz. The filter reduces overall bandwidth down from the width of the entire passband (SR) to the actual width of the signal. This helps eliminate interference from adjacent channels.

The next stage is the detector (demodulator) but as you can see its not there. An interesting feature of IQ signals is that SSB is already demodulated. The left outlet of the [cmul~] object is the upper sideband. All that remains is to amplify signal level and output the audio.

What's Next?

In the next tutorial we'll add a preamp, a band-scope, AGC, and an AM detector. Please send comments and questions to radio@zerokidz.com