

# Software Defined Radio Primer + Project

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# Overview

- What is SDR?
- Why should I care?
- SDR Concepts
- Potential SDR project

## Approach:

This topic is too large and complex to cover comprehensively in a single short presentation. I will try to pick relevant examples to highlight the key elements of SDR

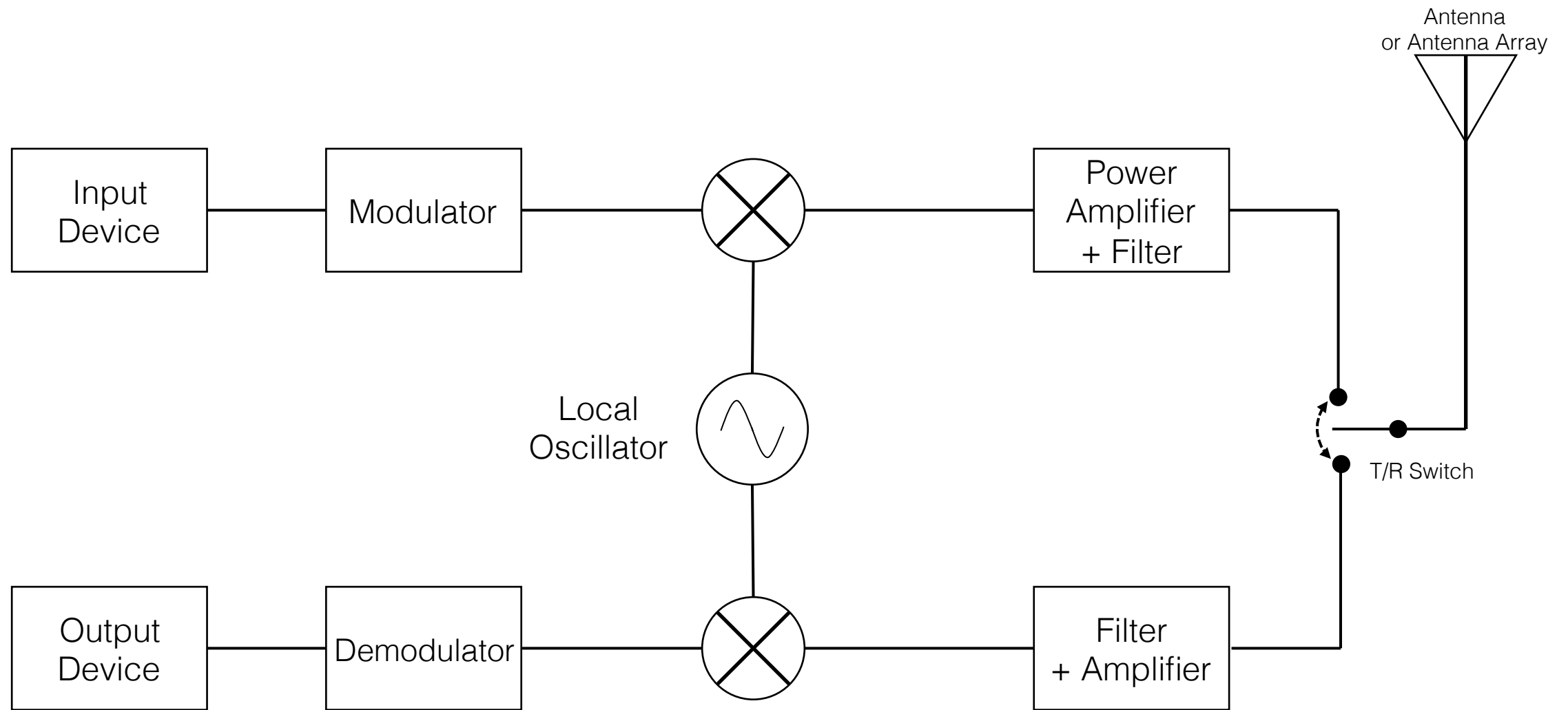
If you have questions...  
ask them!

# What is SDR?

- SDR substitutes software algorithms for functions historically performed by (usually analog) hardware components
- Allows for changing major radio functions with little or no change radio in hardware
  - Eliminates the need for multiple modulation circuits in multi mode transceiver (CW, SSB, AM, FM, RTTY, etc.)
  - Can perform functions difficult or impossible to realize using analog components (certain types of filters, complex modulation techniques, adaptive antenna arrays, etc.)

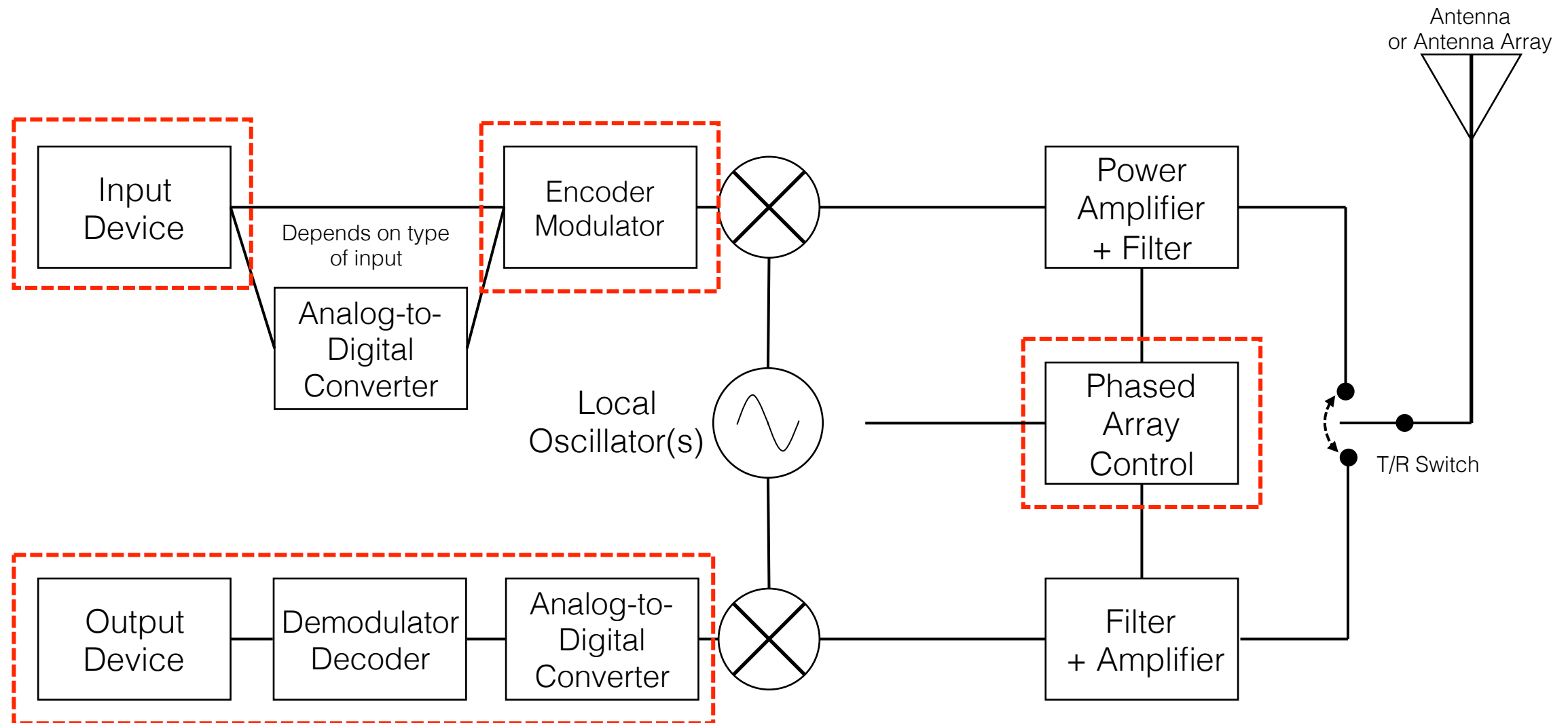
# What is SDR?

## Classic Analog Radio




# What is SDR?

## Software Defined Radio



Note: We sometimes tend to think of SDR only in terms of receivers  
SDR is for transmitters, too!

 Functions which may be performed in software or via computer hardware

# What is SDR?

- Functions typically performed to software (or computer hardware):
  - Analog-to-digital conversion (and often digital-to-analog conversion)
  - Modulation
  - Demodulation
  - Filtering
  - Message coding and decoding
  - Control of phased array antennas (becoming common in new generations of home routers)
  - Radio controls and displays

Can pick and choose which functions to computerize based on operating and design goals

# What is SDR?

- Largely eliminates component tolerance issues when performing signal processing
  - Device bias, phase inaccuracies, filter tuning, etc.
- Very complex math can be performed...it's just math!
- With a fast enough computer, a whole band (or even all of HF) can be acquired and processed in parallel
  - Can even record the data and play back later!
- The ability to store and analyze captured signals allows a computer to perform functions that would require “going back in time” for an analog system
- Can perform statistical analysis to do noise reduction



# What is SDR?

- Functions still typically performed in hardware:
  - RF signal amplification (both transmit and receive)
  - Filtering associated with analog RF amplification (low pass, high pass, band pass, band stop)
  - Audio amplification
  - Power supplies

# Why Should I Care?

- Future
  - SDR is the future of most ham radio equipment
- Cheaper
  - Eventually (if not already) SDRs will be cheaper than classic circuits due to reduced parts counts and fewer circuits requiring post-assembly testing and adjustment
- Flexibility
  - You can do things with SDRs that are difficult or impossible or unaffordable with analog equipment
- Expectations of future hams
  - The next generation of hams will expect more cell phone-like equipment operation (and cell phones already are SDRs...)

# Why Should I Care?

- What solutions are likely to remain analog for a while?
  - Super high performance receivers
  - Super cheap systems
  - Static or targeted-use solutions (e.g., FM repeaters)

# Role of Analog-to-Digital Conversion

- Why?
  - Radio communications always involves some sort of analog signals
  - Computers operate on discrete values
  - Somewhere in the middle analog must be converted to digital (and generally vice versa)
  - Amateur radio involves signals over a wide dynamic range – often greater than A/D converters can adequately manage by themselves
- Cost vs. performance
  - A/D converters used to be expensive
  - Converters covering audio frequencies now pretty cheap
  - High-resolution, high accuracy converters covering high RF frequencies can still be somewhat expensive
  - The processing to handle all that data can also be relatively expensive

You pay for what you get in SDR

# Time Domain vs. Frequency Domain

- In analog systems, we normally do all processing of signals as they exist in the time domain (i.e., as a function of time)
- In SDR, we often process signals in the frequency domain (i.e., as a function of frequency)
  - Conversion from time to frequency and back via a function called the Fast Fourier Transform (FFT)
  - Possible to do frequency domain processing in analog systems, but very complex and expensive (example: Navy F-14 radar)
- Frequency domain processing allows determination of frequency and phase of all signals in the passband

# Coherent vs. Non-Coherent

- Coherent demodulation uses "exact" knowledge of carrier phase to perform demodulation and decoding
  - Allows receiver to integrate (add up) signal over time to improve performance in noise and signal degradation
    - Can receive signals that are substantially weaker than the band noise!
  - Allows the use of exact phase in performing demodulation - mandatory for certain digital communications protocols
  - Harder to do (particularly with analog circuits)
- Non-coherent demodulation does not use exact phase information
  - Doesn't mean phase is ignored altogether
  - Normally uses instantaneous demodulator output (no integration) that results in poorer performance with noise and signal degradation
  - Easier to do

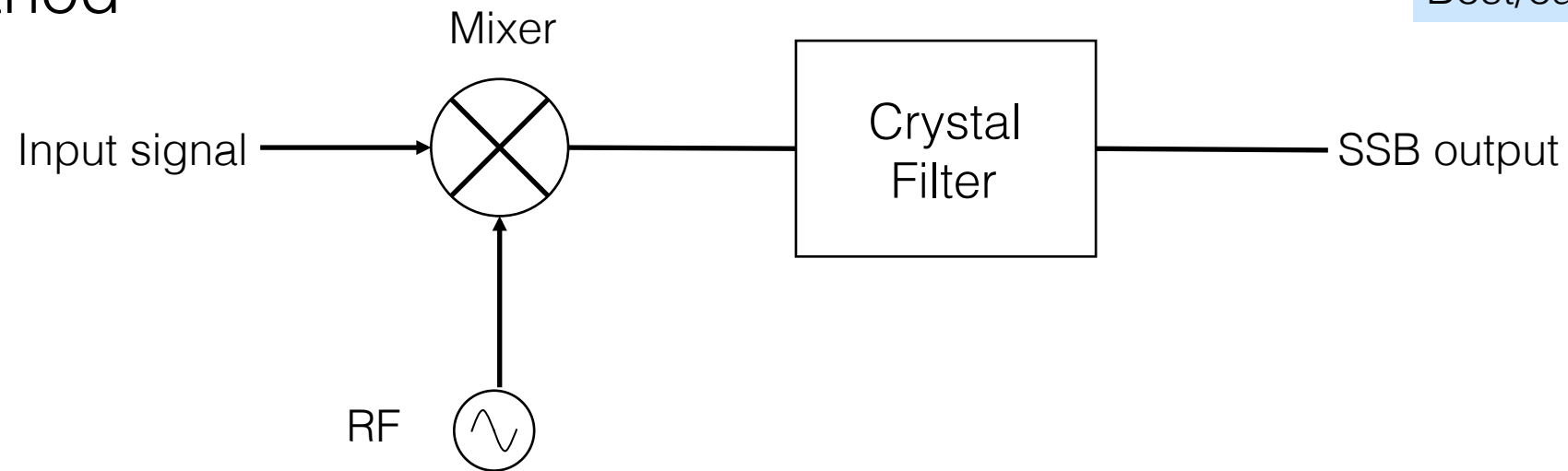
Weak signal work is greatly improved by using coherent detection

# SSB Example

- SSB is created by multiplying the input (usually voice) by an RF signal and then eliminating the unwanted sideband
- Two primary methods of eliminating the unwanted sideband:
  - Filter method
    - Most popular
    - Requires expensive crystal filters
  - Phasing method
    - Generated by phase shifting the input and the RF signals and canceling out the unwanted sideband
    - Hard to do in analog circuitry
    - Pretty easy to do in SDR

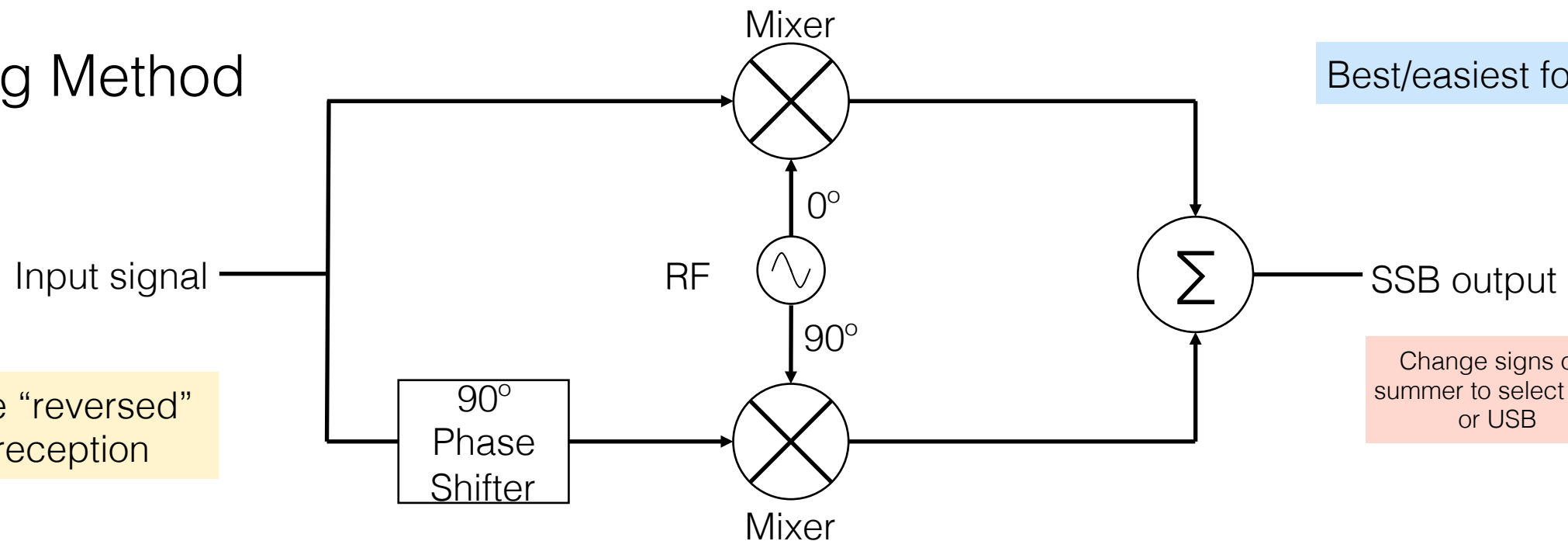
# SSB Block Diagrams

## Filter Method



Best/easiest for analog

## Phasing Method



Best/easiest for SDR

Can be "reversed" for reception

Change signs on summer to select LSB or USB



# SSB Trivia Question

Historically, why was 9 MHz chosen as the “standard” intermediate frequency for HF radios?

# Potential Project

Now let's put this together into a ham radio project

# Project Goals

- Transceiver - not just a receiver like the low-cost "TV dongles" popular today
- QRP - must run on the power available from a standard USB 2.0 port (max 2.5 W total power available to all components in the radio)
- Small physical package – mobile, mobile, mobile (okay, ham shack, too!)
- Predictable, medium/long distance, data communications (even if data rate is slow)
- Software must run unmodified on as many platforms as practical - no drivers other than those built into the operating systems
- No complex software installation, configuration, or operating steps
- Transceiver hardware should NOT require a microcontroller
- Hardware must be under \$100 and preferably under \$50

# Design Choices

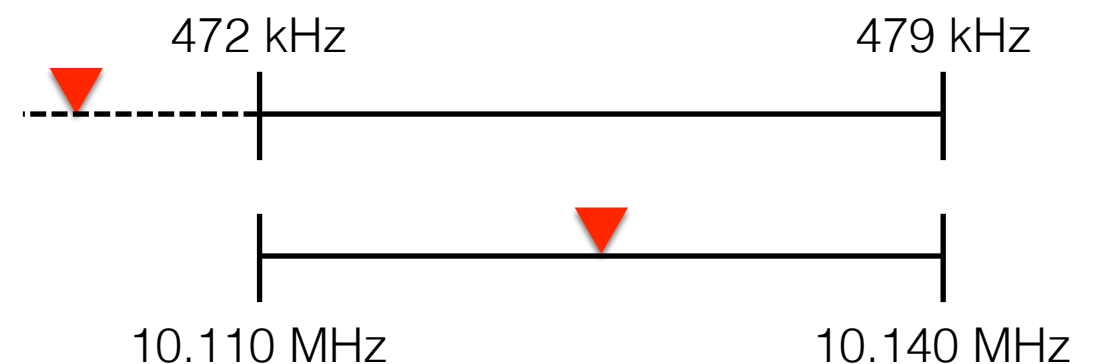
- Hardware

- Assume a single-band, crystal-controlled, direct conversion receiver (and transmitter)
  - Use phasing approach to generate and receive SSB-like signals
  - Tuning controlled by audio frequency (max freq  $< \pm 20\text{kHz}$  or maybe  $< \pm 10\text{kHz}$ )
  - Radio-to-computing device interface exclusively via stereo audio in-out

- Candidate bands

- 630 meters (472 - 479 kHz)
- 30 meters (10.1 - 10.15 MHz)

30 meters is better for first attempt



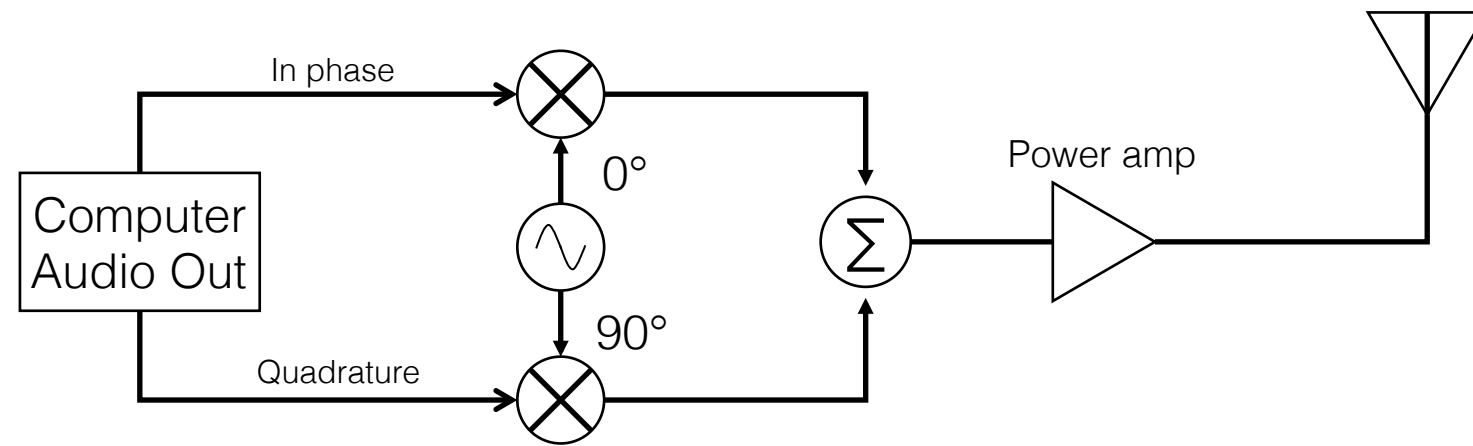
▼ = crystal oscillator frequency

# Design Choices

- Candidate software platforms
  - Web Audio platform using JavaScript in a browser
    - Should support Mac, Windows, Linux, Android, iOS (eventually)
  - Java application using the Java sampled sound API
    - Should support Mac, Windows, Linux, Android (maybe...)

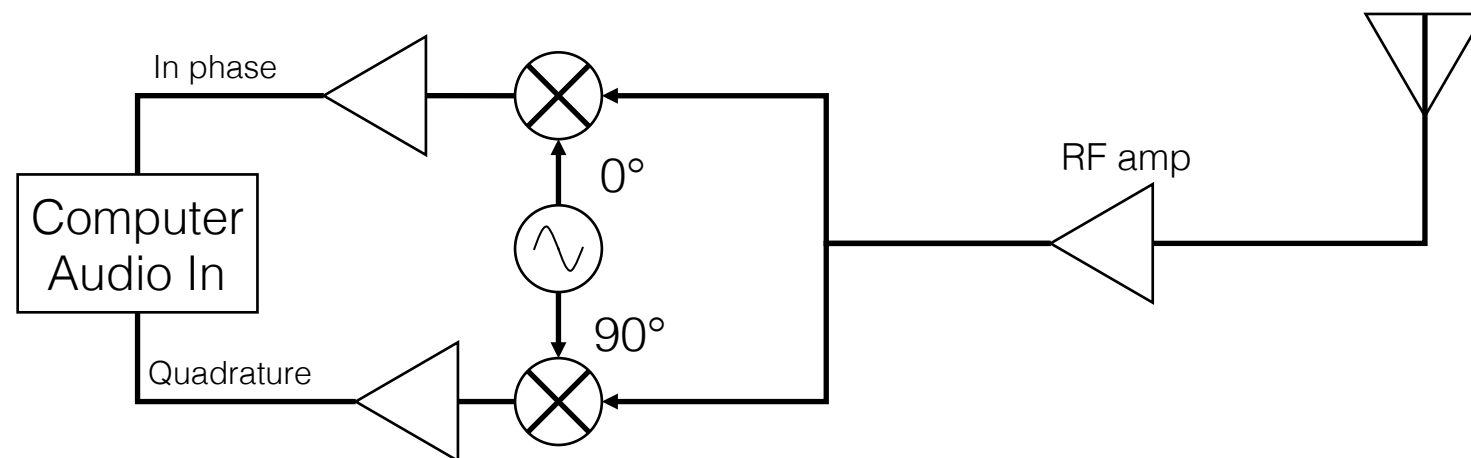
# Simplified Block Diagrams

Transmit



VOX switches between the two circuit configurations  
Reuse oscillator, mixers, and maybe some amplifiers between transmit and receive

Receive



# Candidate Hardware Components

- **Mixers/multipliers**
  - Expensive – AD834 – highly linear, minimal filtering (\$30+ each - ouch!)
  - Mid-range – AD8333 – has two mixers plus oscillator 90° phase shifter built in
  - Cheap – SA612 – used as switching mixer, lots of filtering
- **Small signal amplifiers**
  - A little more expensive – nice video amplifiers, such as AD8055
  - A little cheaper – Mini-Circuits amplifiers – MAR-1, MAR-2, etc.
- **Power amplifier** – really tough due to 5V supply limitations
  - MAX 2602 – intended for cell phone applications (low voltage)
- **Electronic switches** – ADG904, as example
- **Crystal oscillator** – Digi-Key sells programmable oscillators dirt cheap (~\$3 each) that they will pre-program for you
  - Example – Cardinal Components CPP series
  - Open question – will these have too much phase noise???

# Protocol Selection

- Characteristics
  - Single tone at a time - allows for non-linear amplification (higher efficiency)
  - Suitable for weak signals and slow transmission over HF and MF
  - Small bandwidth
  - Must support IQ inputs and outputs
  - Must have thorough specification (too laborious to reverse engineer code)
  - Must be compliant with regulations

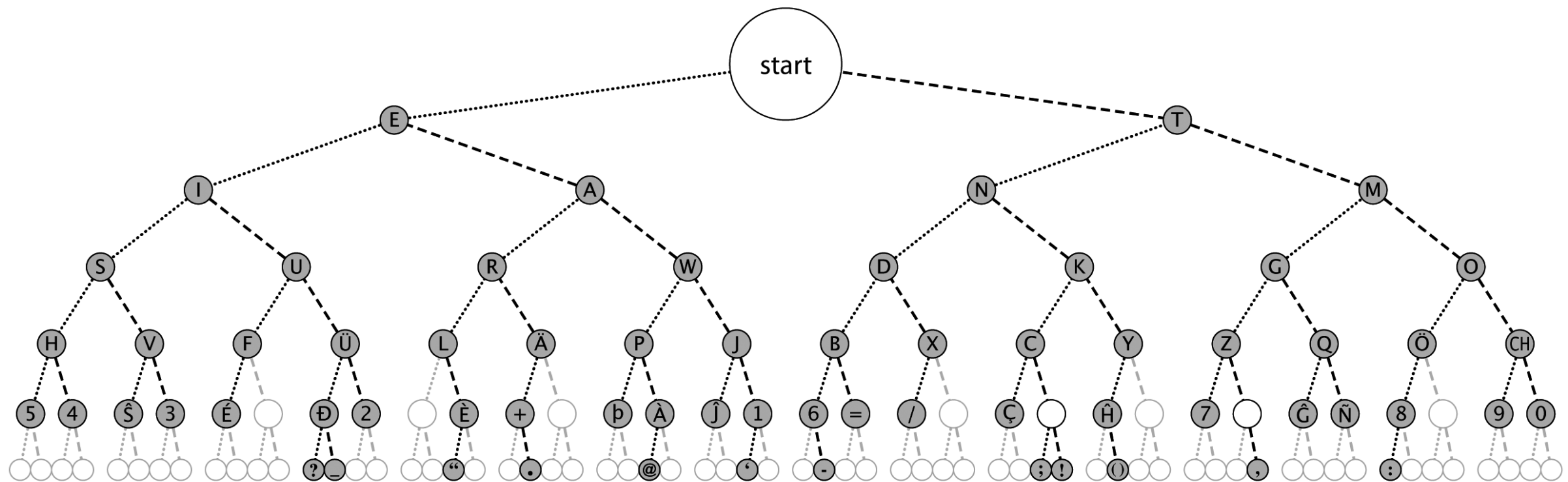


# Protocol Selection

- Strategy
  - Start with Morse code
    - Variable speed
    - Could be as slow as 1 WPM or less at poor signal-to-noise ratios
  - Add option to enhance using two tones rather than on/off
  - Add option to use enhanced character set
    - There are ~60 unused characters at up to 6 dots/dashes
  - Consider other popular weak signal protocols at a later date

# Protocol Selection

## Morse Code Binary Tree



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# Protocol Selection

## Morse Code

- Pros:
  - While not the most efficient protocol, Morse code is surprisingly efficient
  - When using “pure” Morse code, transmissions are legal on virtually all amateur frequencies
  - Morse code extensions allow for a very large symbol base
  - Can easily add error detection – I call it “MorseTOR”
  - Does not rule out human decoding (by ear or “waterfall” display) when signal strength is good (try that with an MFSK signal!)
  - Can use non-linear power amp – better efficiency
  - Simple protocol, well known, no reverse engineering necessary
- Cons:
  - Other weak signal modes have proven to be better performers

# Wrap Up

- SDR is the future of radio...embrace it!
- Prices will come down as competition heats up
- You can use SDR principles to homebrew pretty sophisticated gear on the cheap

# Questions?

I hope you enjoyed the presentation!