

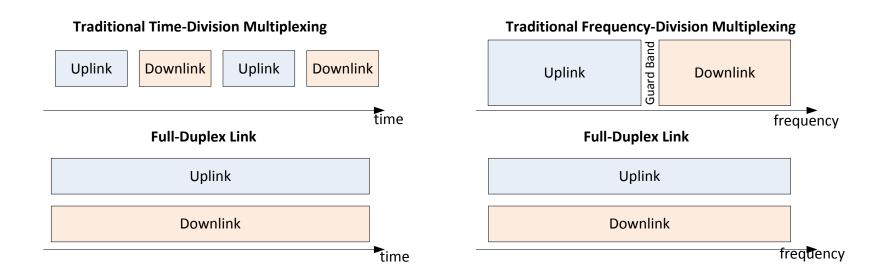
Agenda

- Theory: Full-Duplex
 - What is Full Duplex?
 - The Problem
 - The Solution
- Theory: Digital Cancellation
 - Black Box Model
 - Modeling Nonlinearity
 - Least-squares Problem
 - Algorithms for Solving

- Custom GNU Radio Blocks
 - Implementation
 - Cancellation Performance
 - Throughput Results
- Using USRP Products
 - Timing Synchronization
 - Built-in vs. External Mixers
- Conclusions

What is Full-Duplex Communications?

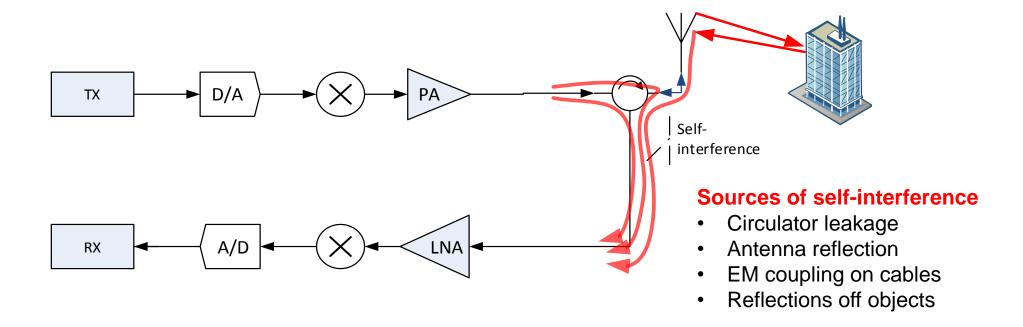
- ... it's not the 'usual definition'
- Transmit and receive on the <u>same antenna</u> on the <u>same frequency</u> at the <u>same time</u>
 - Potentially double (or more) throughput in the same spectrum
 - Considered impossible for typical communications links
 - Multiple classical textbooks explicitly state it can't be done
- Can replace both time- and frequency-multiplexed links



The Problem



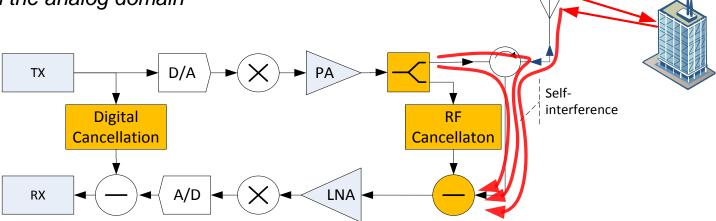
- The problem is self-interference
 - Transmit power swamps the receive power, making it very difficult to detect the desired signal
 - Power difference $10log_{10}(P_{Tx}/P_{Rx})$ depends on the distance between Tx and Rx



The Solution

Very active area of research in the past several years

- The transmitted signal is known at the receiver
 - To some degree, need to account for delay and distortion(s)
- Subtract the transmitted signal from the received signal to remove self-interference
- The devil is in the implementation
 - Transmitted signal power may be > 100 dB above received signal
 - Need very high linearity, very accurate matching and model of distortions
- Two major approaches:
 - 1. Cancel at baseband in the digital domain
 - 2. Cancel at RF in the analog domain



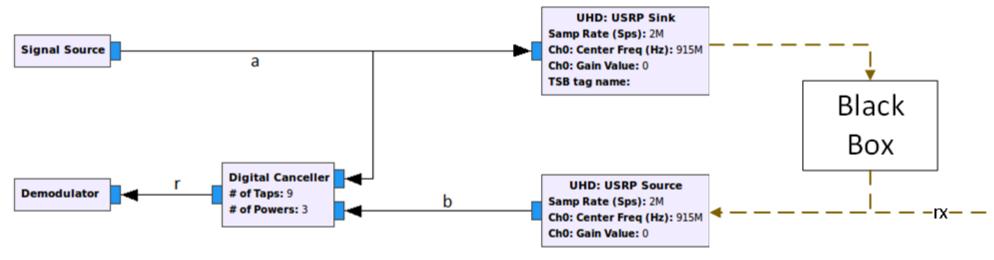
All reported full-duplex systems use both analog and digital cancellation





Digital Cancellation





- Find a model for the black box to minimize *r*.
 - I.e. find \mathcal{F} to minimize $|r| = |b \mathcal{F}(a)|$
- The model must account for:
 - Gain
 - Phase shift
 - Time delay
 - Nonlinear distortion
 - Multipath

Modeling the Black Box

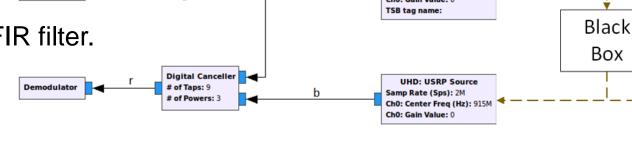
Linear Model

- We model the black box as a non-causal FIR filter.
- Let:

 a_i = transmitted signal, sample i

 b_i = received signal, sample i

 $x_k = FIR$ filter coefficient k



UHD: USRP Sink

Samp Rate (Sps): 2M

- Our goal is to find the filter coefficients that produce outputs (r_i) most similar to the actual received signal.
- Example for a 5-tap filter, for the 2nd received sample:

$$b_2 = x_0 a_0 + x_1 a_1 + x_2 a_2 + x_3 a_3 + x_4 a_4$$

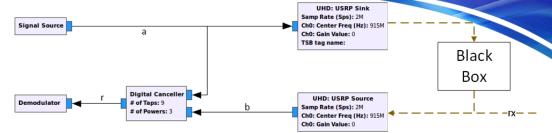
Or, in vector form:

$$\begin{bmatrix} a_0 \ a_1 \ a_2 \ a_3 \ a_4 \end{bmatrix} \begin{bmatrix} x_0 \\ x_1 \\ x_2 \\ x_3 \\ x_4 \end{bmatrix} = b_2$$

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Modeling the Black Box

Nonlinear Distortion



 $[x_{10}]$

 x_{11}

 x_{12}

- We model the black box as a **summation** of FIR filters, each operating on a different (odd) power of the transmitted signal.
- Let:

```
a_i = transmitted signal, sample i
b_i = received signal, sample i
x_{ik} = FIR filter coefficient k for power j
```

Example for 3 powers (1st, 3rd,and 5th), each with 5 taps:

[1st, 3rd, and 5th], each with 5 taps:
$$\begin{bmatrix} a_0 & a_1 & a_2 & a_3 & a_4 & a_0^3 & a_1^3 & a_2^3 & a_3^3 & a_4^3 & a_0^5 & a_1^5 & a_2^5 & a_3^5 & a_4^5 \end{bmatrix} \begin{bmatrix} x_{13} \\ x_{30} \\ x_{31} \\ x_{32} \\ x_{33} \\ x_{34} \end{bmatrix} = b_2$$

Modeling the Black Box

Matrix Representation

• If we wish to work with multiple samples of the received signal at a time, we can express the model in matrix form as follows:

$$\begin{bmatrix} a_0 & a_1 & a_2 & a_3 & a_4 & a_0^3 & a_1^3 & a_2^3 & a_3^3 & a_4^3 & a_0^5 & a_1^5 & a_2^5 & a_3^5 & a_4^5 \\ a_1 & a_2 & a_3 & a_4 & a_5 & a_1^3 & a_2^3 & a_3^3 & a_4^3 & a_5^3 & a_1^5 & a_2^5 & a_3^5 & a_4^5 & a_5^5 \\ a_2 & a_3 & a_4 & a_5 & a_6 & a_2^3 & a_3^3 & a_4^3 & a_5^3 & a_6^3 & a_2^5 & a_3^5 & a_4^5 & a_5^5 & a_6^5 \\ a_3 & a_4 & a_5 & a_6 & a_7 & a_3^3 & a_4^3 & a_5^3 & a_6^3 & a_7^3 & a_3^3 & a_4^5 & a_5^5 & a_6^5 & a_7^5 \\ a_4 & a_5 & a_6 & a_7 & a_8 & a_4^3 & a_5^3 & a_6^3 & a_7^3 & a_8^3 & a_5^5 & a_5^5 & a_5^5 & a_5^5 & a_5^5 \\ a_5 & a_6 & a_7 & a_8 & a_9 & a_5^3 & a_6^3 & a_7^3 & a_8^3 & a_9^3 & a_5^5 & a_6^5 & a_7^5 & a_8^5 \\ a_5 & a_6 & a_7 & a_8 & a_9 & a_5^3 & a_6^3 & a_7^3 & a_8^3 & a_9^3 & a_5^5 & a_6^5 & a_7^5 & a_8^5 \\ a_5 & a_6 & a_7 & a_8 & a_9 & a_5^3 & a_6^3 & a_7^3 & a_8^3 & a_9^3 & a_5^5 & a_6^5 & a_7^5 & a_8^5 \\ a_5 & a_6 & a_7 & a_8 & a_9 & a_5^3 & a_6^3 & a_7^3 & a_8^3 & a_9^3 & a_5^5 & a_6^5 & a_7^5 & a_8^5 \\ a_5 & a_6 & a_7 & a_8 & a_9 & a_5^3 & a_6^3 & a_7^3 & a_8^3 & a_9^3 & a_5^5 & a_6^5 & a_7^5 & a_8^5 \\ a_5 & a_6 & a_7 & a_8 & a_9 & a_5^3 & a_6^3 & a_7^3 & a_8^3 & a_9^3 & a_5^5 & a_6^5 & a_7^5 & a_8^5 \\ a_5 & a_6 & a_7 & a_8 & a_9 & a_5^3 & a_6^3 & a_7^3 & a_8^3 & a_9^3 & a_5^5 & a_6^5 & a_7^5 & a_8^5 \\ a_7 & a_8 & a_9 & a_5^3 & a_6^3 & a_7^3 & a_8^3 & a_9^3 & a_5^5 & a_6^5 & a_7^5 & a_8^5 \\ a_7 & a_8 & a_9 & a_5^3 & a_6^3 & a_7^3 & a_8^3 & a_9^3 & a_5^5 & a_6^5 & a_7^5 & a_8^5 \\ a_7 & a_8 & a_9 & a_5^3 & a_6^3 & a_7^3 & a_8^3 & a_9^3 & a_5^5 & a_6^5 & a_7^5 & a_8^5 \\ a_7 & a_8 & a_9 & a_5^3 & a_6^3 & a_7^3 & a_8^3 & a_9^3 & a_5^5 & a_6^5 & a_7^5 & a_8^5 \\ a_7 & a_8 & a_9 & a_5^3 & a_6^3 & a_7^3 & a_8^3 & a_9^3 & a_5^5 & a_6^5 & a_7^5 & a_8^5 \\ a_7 & a_8 & a_9 & a_5^5 & a_6^5 & a_7^5 & a_8^5 & a_9^5 \\ a_7 & a_8 & a_9 & a_5^5 & a_6^5 & a_7^5 & a_8^5 \\ a_7 & a_8 & a_9 & a_5^5 & a_6^5 & a_7^5 & a_8^5 \\ a_7 & a_8 & a_9 & a_7^5 & a_8^5 & a_7^5 & a_8^5 \\ a_7 & a_8 & a_9 & a_7^5 & a_8^5 & a_7^5 & a_8^5 \\ a_7 & a_8 & a_9 & a_7^5 & a_8^5 & a_7^5 \\ a_8 & a_8 & a_9 & a_8^5 & a_$$

$$\begin{bmatrix} x_{10} \\ x_{11} \\ x_{12} \\ x_{13} \\ x_{14} \\ x_{30} \\ x_{31} \\ x_{32} \\ x_{33} \\ x_{34} \\ x_{50} \\ x_{51} \\ x_{52} \\ x_{53} \\ x_{54} \end{bmatrix} = \vec{b}_{2}$$

$$\begin{bmatrix} b_{2} \\ b_{3} \\ b_{4} \\ b_{5} \\ b_{6} \\ b_{7} \\ \vdots \end{bmatrix}$$

$$\vec{x} = \vec{b}$$

We are looking for the value of \vec{x} that minimizes $\|A\vec{x} - \vec{b}\|$

Algorithms



For Solving $A\vec{x} = \vec{b}$

Block-based algorithms operate on the entire matrix A, effectively computing $\vec{x} = A^+ \vec{b}$.

- Singular value decomposition (SVD)
- QR decomposition
- Cholesky decomposition
- Conjugate gradient method

Adaptive algorithms operate on one row of A at a time, adjusting the value of \vec{x} each iteration.

- Least mean squares (LMS)
- Recursive least squares (RLS)

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We implemented a GNU Radio block for each algorithm in green.

Note



These GNU Radio blocks are not just for full duplex!

 They can be used to solve any problem where it is desirable to cancel a known signal that may have undergone linear and/or nonlinear distortion.

Performance Optimization



- For high throughput performance, we leverage the Intel libraries:
 - **IPP**: Intel Performance Primitives:
 - Contains signal processing routines
 - Optimized using Streaming SIMD Extensions
 - MKL: Math Kernel Library
 - Contains optimized functions for vector and matrix math
 - API is compatible with BLAS and LAPACK functions

• We further increase throughput by dividing computationally-intensive work between multiple threads.

Implementation (Pseudocode)

SVD Block

```
int svd_canceller_cc_impl::general_work(
   int noutput_items, gr_vector_int &ninput_items,
   gr_vector_const_void_star &input_items, gr_vector_void_star &output_items)
   gr_complex* ref = input_signals[0]; // ref = transmitted signal
   gr complex* b = input signals[1]; // b = received signal
   gr_complex* out = output_signals[0];
   construct_A_matrix(ref, A);
   cgelsd(A, b, x, ...);
                                          // solve Ax = b for x (using SVD)
                                          // residual = b - Ax
   cgemv(A, x, b, out, ...);
   consume_each(block_size);
   return block size;
```

Implementation (Pseudocode)

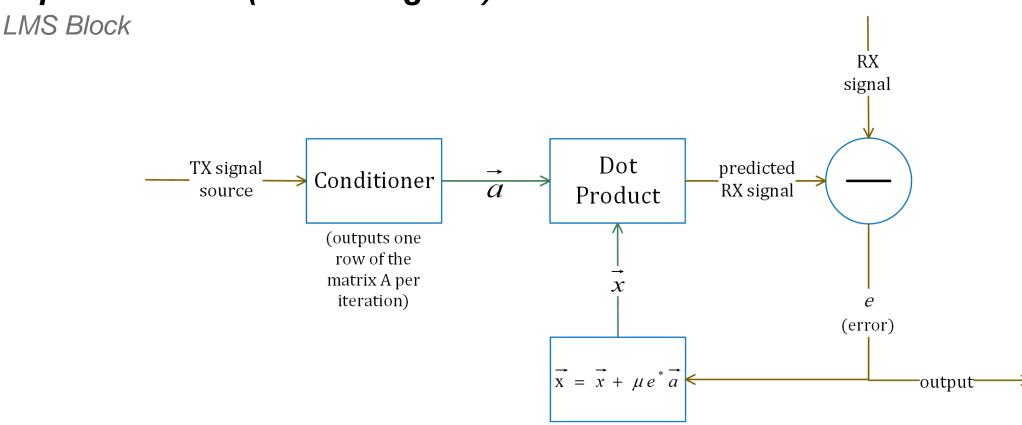
QR Decomposition Block

```
int qr_canceller_cc_impl::general_work(
    int noutput_items, gr_vector_int &ninput_items,
    gr_vector_const_void_star &input_items, gr_vector_void_star &output_items)
   gr_complex* ref = input_signals[0]; // ref = transmitted signal
    gr complex* b = input signals[1]; // b = received signal
    gr_complex* out = output_signals[0];
    construct_A_matrix(ref, A);
                                           // perform QR factorization on A
    cgeqrf(A, temp, ...);
   cungqr(temp, Q, ...);
                                           // compute Q explicitly
    cgemv(Q, b, Qb, ...);
                                           // compute Q^Tb
                                           // residual = b - Q Q^T b
    cgemv(Q, Qb, b, out, ...);
    consume_each(block_size);
    return block size;
```



Implementation (Block Diagram)





- We optimize this block for throughput by:
 - Using multiple threads, each of which processes a chunk of the input
 - Calculating updates to the coefficients \vec{x} every N samples (instead of every sample)
 - Averaging the threads' values of \vec{x} when they synchronize

Results and Performance

Cancellation

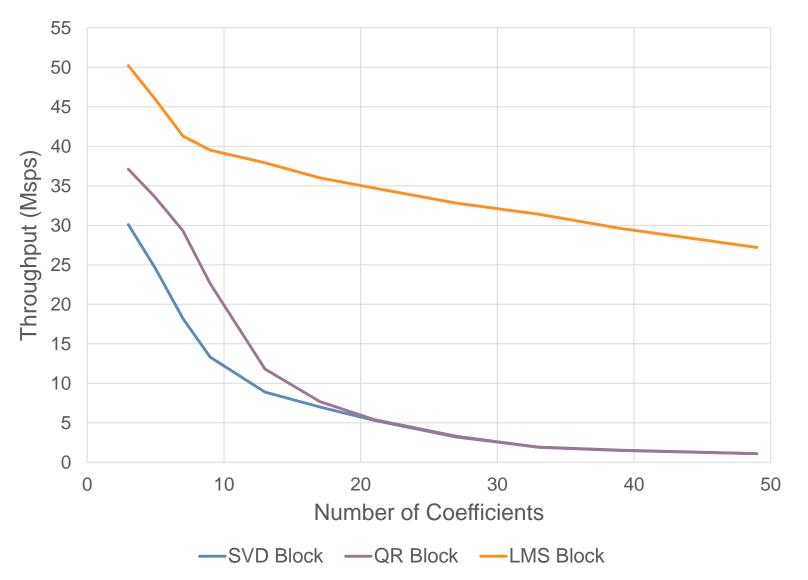


Test conditions:

- Digital simulation
- 13-tap FIR filter
- 1 power (linear only)
- SVD/QR block size: 512
- LMS step size: 0.01

Results and Performance

Throughput



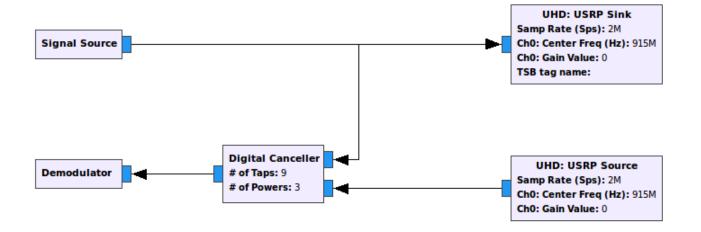
Test conditions:

- Intel Xeon X5660 (2.80GHz)
- 8 threads

• If you decide to use USRPs in a full-duplex radio system, here are some tips ...



Timing Synchronization



- For this flowgraph to work, the USRP sink and source blocks must start streaming at the same time.
- These blocks provide functions for timing synchronization:
 - set_time_now()
 - set_time_next_pps()
 - set_start_time()
- We can edit the GRC-generated Python code to call these functions ...
 - But if we make a change to the flowgraph and regenerate, we have to do it again
 - There must be a better way ...

Function Caller Block



 Wouldn't it be nice if there was a GRC block that allowed us to embed arbitrary function calls in the generated code, that execute before the flowgraph starts?

We created one:

Function Caller Function Caller ID: caller3 ID: caller2 Function Belongs to: Block Function Belongs to: Block Block ID: usrp sink Block ID: usrp source Function Name: set time now Function Name: set time now Function Args: 0 Function Args: 0 **Function Caller Function Caller** ID: caller0 ID: caller1 Function Belongs to: Block Function Belongs to: Block Block ID: usrp sink Block ID: usrp source Function Name: set start time Function Name: set start time Function Args: 1 Function Args: 1

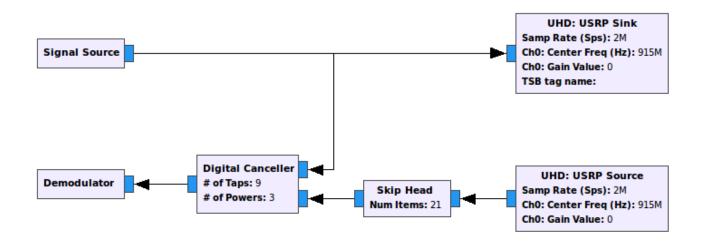
Generated Python code:

```
val = self.usrp_source.set_time_now(0)
...
val = self.usrp_sink.set_time_now(0)
...
val = self.usrp_source.set_start_time(1)
...
val = self.usrp_sink.set_start_time(1)
```

Largely based on the built-in Function Probe block

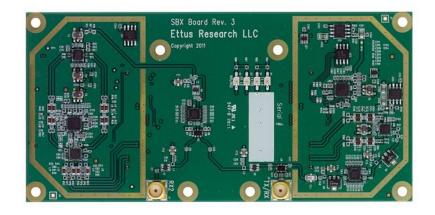
Timing Synchronization

- Even after synchronizing the USRP source and sink, there is still a timing offset:
 - Deterministic and repeatable
 - Appears to be sample-rate dependent
 - On the order of 10-50 samples
- This can be remedied with a skip head block:

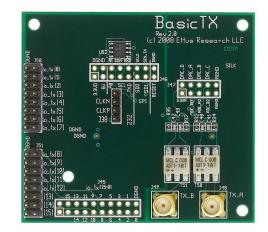


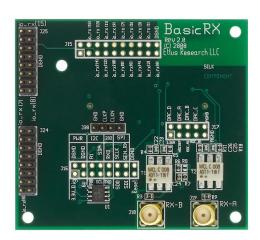
Built-in vs. External Mixers

- Most USRP daughterboards contain LO generators and mixers to convert between baseband/IF and RF.
- e.g. SBX, UBX



- Some daughterboards contain no LO or mixer but operate at baseband/IF.
- e.g. BasicTX, BasicRX

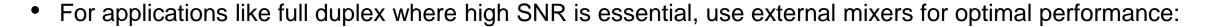




Source: https://www.ettus.com/product/category/Daughterboards

 These boards can be used with external mixers to produce RF.

Built-in vs. External Mixers



Digital Cancellation (dB)		Transmitter Setup	
		BasicTX + External Up- converter	SBX
Receiver Setup	BasicRX + External Down- converter	54.4	44.6
	SBX	39.2	38.0

Test conditions:

- USRP X310
- Analog loopback

Summary and Conclusions



- It is feasible to implement a full-duplex radio system using GNU Radio.
- By using Intel libraries and multi-threading, we can support bandwidths in the tens of MHz.
- If parallelized, adaptive algorithms like LMS provide higher throughput with minimal cost to cancellation.
- Digital cancellation blocks can be applied to any scenario involving suppression of a known signal.

Questions?

