# RITSAR

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# April 28, 2015

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### 1 Introduction

Below is a description of each of the functions contained in the RITSAR package. For installation instructions, reference the README file.

Each of the sections below represent a module contained within the RITSAR package and each subsection details how to use each function. The format for the title of each subsection is "function(parameters)." This is followed by a short description of what the function does. Next, each parameter is defined. Finally, a description of what the function returns is given.

As an example, lets say you wanted to simulate your own phase history. The function required for this is simulate\_phs located in the phsTools module. Before this function can be used, the required parameters need to be defined. The descriptions of each required parameter can be found in subsection 2.1. The first parameter that needs to be defined is "platform", which is a Python dictionary. A description of each key that goes into platform is also given in subsection 2.1. Once platform is defined, the user then creates an array specifying the locations of each point reflector in the scene as well as the corresponding amplitudes. After all this is done the user can then create their own phase history, stored in the phs variable, by typing the following into the Python console:

phs = ritsar.phsTools.simulate\_phs(platform, points, amplitudes)

For a full end-to-end demonstration, "main.py" in the ./examples folder can be run. The default setup processes an X-band collect using the polar format algorithm. The final lines of main.py can be commented/uncommented to use the backprojection algorithm. For a demonstration of the omega-k algorithm, OpenCV will be required. You will most likely also want to use a UHF setup. To do this, simply change SARplatform to SARplatformUHF in the demo and comment/un-comment the bottom lines of the file as necessary. The platform dictionary keys defined in SARplatform and SARplatformUHF were taken from Carrera [1].

# 2 phsTools

## 2.1 simulate\_phs(platform, points, amplitudes)

This function is a <u>basic</u> phase history simulator. It takes a list of target locations and amplitudes and saves the demodulated signal to './phase\_history.npy'. It also outputs the signal to the function caller. The amplitude of the received signal for a perfect point reflector is assumed to be one arbitrary unit. A uniform beam pattern is assumed and  $1/r^4$  attenuation is not modeled.

### Parameters:

• platform : dictionary dictionary that has the following keys defined:

nsamples : int number of fast time samples

npulses: int number of pulses
chirprate: float chirprate in Hz/s

- **f\_0** : float carrier frequency in Hz

- t : float array [nsamples] <u>demodulated</u> fast-time <u>centered at t=0</u>. The array should go from  $(-nsamples/2, nsamples/2) \times \Delta t$  where,  $\Delta t = (A/D \text{ sampling rate})^{-1}$ 

pos: float array [npulses x 3]
 (x,y,z) coordinates of platform, referenced to scene center, which is located at (0,0,0)

• **points**: float array [npoints x 3] locations of each point referenced to scene center

• amplitudes: float array [npoints] reflectance of each point. May be a complex number.

#### Returns:

• **phase\_history** : complex array [npulses x nsamples] resultant phase history

# 2.2 RVP\_correct(phs, platform)

Corrects Residual Video Phase using the formulation in Carrera Appendix C.

### Parameters:

- **phs**: complex array [npulses x nsamples] complex phase history to be corrected for residual video phase
- platform : dictionary dictionary that has the following keys defined:
  - nsamples : int number of fast time samples
  - npulses: int number of pulses
    chirprate: float chirprate in Hz/s
  - **delta\_r**: float spacing between fast time samples. Defined here as  $\frac{c}{2B_{IF}}$  where,  $B_{IF}$  is the bandwidth of the intermediate frequency (de-chirped) signal.  $B_{IF}$  is simply  $(t_{max} - t_{min}) \times chirprate$  where t is the demodulated fast time, as defined in subsection 2.1.

#### Returns:

• **phase\_history**: complex array [npulses x nsamples] resultant phase history, corrected for residual video phase

# 2.3 phs\_to\_const\_ref(phs, platform, upchirp)

This program converts a phase history that was demodulated using a pulse dependant range to scene center to a phase history that is demodulated using a fixed reference. The fixed reference is defined as the minimum range to scene center.

#### Parameters:

- **phs**: complex array [npulses x nsamples] complex phase history that was compensated to a point
- platform: dictionary dictionary that has the following keys defined:
  - **npulses**: int number of pulses
  - **chirprate** : float chirprate in Hz/s
  - **f\_0** : float carrier frequency in Hz
  - **t**: float array [nsamples] <u>demodulated</u> fast-time <u>centered at t=0</u>. The array should go from  $(-nsamples/2, nsamples/2) \times \Delta t$  where,  $\Delta t = (A/D \text{ sampling rate})^{-1}$
  - **pos**: float array [npulses x 3] (x,y,z) coordinates of platform, referenced to scene center, which is located at (0,0,0)

### Returns:

• **phase\_history**: complex array [npulses x nsamples] resultant phase history, compensated to a line

# 3 imgTools

In this section the img\_plane dictionary is introduced. A sample img\_plane dictionary generator is provided in sarit/examples/dictionaries.

## 3.1 polar\_format(phs, platform, img\_plane, taylor)

This is the Polar Format algorithm. The phase history data as well as platform and image plane dictionaries are taken as inputs.

The phase history data is collected on a two-dimensional surface in k-space. For each pulse, a strip of this surface is collected. The first step in this program is to project each strip onto the (ku,kv) plane defined by the normal vector contained in the image plane dictionary. This will result in data that is unevenly spaced in (ku,kv). This unevenly spaced data is interpolated onto an evenly spaced (ku,kv) grid defined in the image plane dictionary. The interpolation is done along the radial direction first, then along the along-track direction. Further details of this method are given in both the Jakowitz and Carrera texts.

#### Parameters:

- **phs** : complex array [npulses x nsamples] complex phase history
- platform : dictionary dictionary that has the following keys defined:
  - npulses : int number of pulses
  - f\_0 : float carrier frequency in Hz
  - **pos**: float array [npulses x 3] (x,y,z) coordinates of platform, referenced to scene center, which is located at (0,0,0)
  - **k\_r**: float array [nsamples] spatial frequencies corresponding to each fast time return. For a linear FM signal, this is given by  $\frac{4\pi}{c}(f_0 + chirprate \times t)$
  - R\_c : float vector to scene center at aperture center
- img\_plane : dictionary dictionary that has the following keys defined:
  - n\_hat : float array [3]unit normal vector to image plane
  - $-\mathbf{k}_{-}\mathbf{u}$ : float array [size(u)] image plane spatial frequencies along u axis. u is defined to lie along the slant range vector, projected onto the image plane.
  - $k_v$ : float array [size(v)] image plane spatial frequencies along v axis. v is defined to lie perpendicular to the slant range vector in the image plane.
- taylor: int attenuation factor for taylor window in dB. defaults to 43 dB.

### Returns:

• img : complex array [npulses x nsamples] processed image

## 3.2 backprojection(phs, platform, img\_plane, taylor, upsample)

This is the Backprojection algorithm. The phase history data as well as platform and image plane dictionaries are taken as inputs. The (x,y,z) locations of each pixel are required, as well as the size of the final image (interpreted as  $[size(v) \times size(u)]$ ).

#### Parameters:

- **phs** : complex array [npulses x nsamples] complex phase history
- platform: dictionary dictionary that has the following keys defined:
  - nsamples : int number of fast time samples
  - npulses : int number of pulses
  - **pos** : float array [npulses x 3] (x,y,z) coordinates of platform, referenced to scene center, which is located at (0,0,0)
  - **k\_r**: float array [nsamples] spatial frequencies corresponding to each fast time return. For a linear FM signal, this is given by  $\frac{4\pi}{c}(f_0 + chirprate \times t)$
  - **delta\_r**: float spacing between fast time samples. Defined here as  $\frac{c}{2B_{IF}}$  where,  $B_{IF}$  is the bandwidth of the intermediate frequency (de-chirped) signal.  $B_{IF}$  is simply  $(t_{max} - t_{min}) \times chirprate$  where t is the demodulated fast time, as defined in subsection 2.1.
- img\_plane : dictionary dictionary that has the following keys defined:
  - u: float array [len(u)]
     for the backprojection algorithm, u is essentially a placeholder that defines one of the dimmensions of the output image. Only the length of u matters, not its contents.
  - $\mathbf{v}$ : float array [len(v)] for the backprojection algorithm, v is essentially a placeholder that defines one of the dimmensions of the output image. Only the length of v matters, not its contents.
  - **pixel\_locs**: float array [3 x n\_pixels] flattened array specifying the 3-dimensional location of each pixel in the rectangular output array whose shape is [len(u) x len(v)]. Consequently,  $n_pixels = len(u) \times len(v)$ . The locations do not have to lie in a plane and can be specified to an arbitrary surface that matches the local terrain. A rectangular output array with the location of each pixel specified is the only constraint.
- taylor : int attenuation factor for taylor window in dB. defaults to 43 dB.
- upsample : int upsampling factor, default is 6.

### Returns:

• img : complex array [npulses x nsamples] processed image

# 3.3 omega-k(phs, platform, taylor, upsample)

This is an omega-k algorithm based off of the algorithm prescribed in the Carrera text. Only the phase history and platform files are taken as inputs, an img\_plane dictionary is not required.

The input phase history needs to have been demodulated to a fixed reference. If demodulated to scene center, the included phs\_const\_ref file can do the conversion for you. A straight line flight path is also assumed.

The first step in the algorithm is to perform a 1D FT along azimuth. A matched filter is applied to the resultant data to perfectly compensate the range curvature of all scatterers having minimum range R\_s. The default setting for R\_s is the minimum range of scene center. To correct the range curvature for other scatterers, the data is mapped onto a new grid.

#### Parameters:

- **phs** : complex array [npulses x nsamples] complex phase history
- platform : dictionary dictionary that has the following keys defined:
  - nsamples : int number of fast time samples
  - npulses : int number of pulses
  - pos: float array [npulses x 3]
     (x,y,z) coordinates of platform, referenced to scene center, which is located at (0,0,0)
  - $\mathbf{k}_{-}\mathbf{r}$ : float array [nsamples] spatial frequencies of demodulated signal corresponding to each fast time return. For a linear FM signal, this is given by  $\frac{4\pi}{c}(f_0 + chirprate \times t)$
  - k\_y: float
     spatial frequencies along flight path. A straight line flight path (or phase history motion compensated to a straight line flight path) are required.
- taylor: int attenuation factor for taylor window in dB. defaults to 43 dB.
- upsample : int upsampling factor, default is 6.

#### Returns:

• img : complex array [npulses x nsamples] processed image

# 3.4 img\_plane\_dict(platform, res\_factor, n\_hat, aspect, upsample)

This function defines the image plane parameters. The user specifies the image resolution using the res\_factor. A res\_factor of 1 yields a (u,v) image plane whose pixels are sized at the theoretical resolution limit of the system (derived using delta\_r which in turn was derived using the bandwidth. The user can also adjust the aspect of the image grid. This defaults to nsamples/npulses.

'n hat' is a user specified value that defines the image plane orientation w.r.t. to the nominal ground plane.

#### Parameters:

• platform : dictionary

dictionary that has the following keys defined:

- nsamples : int

number of fast time samples

npulses : int number of pulses

 $-\mathbf{R}_{-}\mathbf{c}$ : float array [3]

vector to scene center at aperture center

- **delta\_r**: float array [nsamples] spacing between fast time samples. Defined here as  $\frac{c}{2B_{IF}}$  where,  $B_{IF}$  is the bandwidth of the intermediate frequency (de-chirped) signal.  $B_{IF}$  is simply  $(t_{max} t_{min}) \times chirprate$  where t is the demodulated fast time, as defined in subsection 2.1.
- res\_factor : float

image resolution in units of theoretical resolution size. This defaults to 1.0.

- n\_hat: float array [3] unit normal vector to image plane. Defaults to (0,0,1).
- aspect : float

aspect ratio of range to cross range. Determines cross-range resolution by multiplying res\_factor by aspect. Defaults to nsamples/npulses

• upsamples : bool

Option to upsample number of pixels. The number of pixels in the range direction is set to a power of 2 that is one power of 2 higher than nsamples. If nsamples is already a power of 2, the number of range pixels is just set to nsamples. The same goes for npulses and number of cross-range pixels.

#### Returns:

 $\bullet$  img\_plane : dictionary

dictionary that has the following keys defined:

- n\_hat : float array [3]
   unit normal vector to image plane. Defaults to (0,0,1).
- **u**: float array [len(u)] pixel locations in range direction. u is defined to lie along the slant range vector, projected onto the image plane.
- v: float array [len(v)]
   pixel locations in cross-range direction. v is defined to lie perpendicular to the slant range vector in the image plane.
- du : float array [len(u)]pixel spacing in range direction
- dv : float array [len(v)]pixel spacing in cross-range direction

- k\_u: float array [size(u)]
   image plane spatial frequencies along u axis. u is defined to lie along the slant range vector,
   projected onto the image plane.
- $-\mathbf{k}_{-}\mathbf{v}$ : float array [size(v)] image plane spatial frequencies along v axis. v is defined to lie perpendicular to the slant range vector in the image plane.
- **pixel\_locs**: float array [3 x n\_pixels] flattened array specifying the 3-dimensional location of each pixel in the rectangular output array whose shape is [len(u) x len(v)]. Consequently,  $n_pixels = len(u) \times len(v)$ . The locations do not have to lie in a plane and can be specified to an arbitrary surface that matches the local terrain. A rectangular output array with the location of each pixel specified is the only constraint.

## 3.5 autoFocus(img, win, win\_params)

This program autofocuses an image using the Phase Gradient Algorithm. If the parameter win is set to auto, an adaptive window is used. Otherwise, the user sets win to 0 and defines win\_params. The first element of win\_params is the starting windows size. The second element is the factor by which to reduce it by for each iteration. This algorithm is based off [2].

### Parameters:

- img : complex array [npulses x nsamples] complex image
- win : string if 'auto', an adaptive window is used. Otherwise, win\_params is used to create the window. Default is 'auto.'
- win\_params: list

  The first element of win\_params is the starting windows size. The second element is the factor by which to reduce it by for each iteration.

#### Returns:

• img\_af : complex array [npulses x nsamples] autofocused image

# 4 phsRead

This section describes the RITSAR interfaces to various datasets. In a nutshell, the user supplies the directory containing the phase history and auxillary data. The functions below essentially perform 3 tasks:

- 1. read in the phase history and export as a NumPy array
- 2. read in the auxillary data
- 3. place auxillary data in a platform dictionary and export

As shown in the functions below, code can be written to handle many different data formats including .envi, .mat, .xml, and binary. As of now, RITSAR can interface with DIRSIG and AFRL Gotcha data. A function is also provided that processes a Sandia data set. These functions serve mostly to show that data in almost any format can be processed with the RITSAR toolset provided code is written that performs the three tasks listed above for a given format.

# 4.1 DIRSIG(directory)

This function reads in the DIRSIG xml data as well as the envi header file from the user supplied directory. The phs and a Python dictionary compatible with RITSAR are returned to the function caller. More information on DIRSIG's SAR capability can be found here: http://www.dirsig.org/docs/new/radar.html

#### Parameters:

• directory: string string containing directory of DIRSIG data to include the xml and envi header files.

#### Returns:

- **phase\_history**: complex array [npulses x nsamples] the complex phase history
- platform : dictionary dictionary that has the following keys defined:
  - **f**\_**0** : float carrier frequency in Hz
  - **chirprate** : float chirprate in Hz/s
  - nsamples : int

number of fast time samples

- npulses : int number of pulses
- B: float bandwidth of transmitted signal
- B\_IF: float bandwidth of intermediate frequency, demodulated signal.
- **delta\_r**: float spacing between fast time samples. Defined here as  $\frac{c}{2B_{IF}}$  where,  $B_{IF}$  is the bandwidth of the intermediate frequency (de-chirped) signal.  $B_{IF}$  is simply  $(t_{max} - t_{min}) \times chirprate$  where t is the demodulated fast time, as defined in subsection 2.1.
- delta\_t : float time between fast time samples. Defined as (A/D sampling rate)  $^{-1}$
- vp : float velocity of platform in m/s

- pos: float array [npulses x 3]
   (x,y,z) coordinates of platform, referenced to scene center, which is located at (0,0,0)
- R\_c: float array [3]
   vector to scene center at aperture center
- t : float array [nsamples] <u>demodulated</u> fast-time <u>centered at t=0</u>. The array should go from  $(-nsamples/2, nsamples/2) \times \Delta t$  where,  $\Delta t = (A/D \text{ sampling rate})^{-1}$
- **k\_r**: float array [nsamples] spatial frequencies of demodulated signal corresponding to each fast time return. For a linear FM signal, this is given by  $\frac{4\pi}{c}(f_0 + chirprate \times t)$
- k\_y: float
   spatial frequencies along flight path. A straight line flight path (or phase history motion compensated to a straight line flight path) are required.
- metadata : string array xml metadata

## 4.2 AFRL(directory, pol, start\_az, n\_az=3)

This function reads in the AFRL Gotcha \*.mat files from the user supplied directory and exports both the phase history and a Python dictionary compatible with RITSAR. The directory is specified by inputting the top-level location which contains all pass data and specifying the polarization, starting azimuth, and number of degrees of azimuth to process.

#### Parameters:

- directory : string string containing directory that contains all the data for a given pass. For example 'AFRL\_data/pass1'
- pol: string polarization to be processed. For example, 'HH'.
- start\_az : int starting azimuth (0 to 360)
- n\_az : int number of azimuth files to process. Default is 3.

#### Returns:

- **phase\_history**: complex array [npulses x nsamples] the complex phase history
- platform: dictionary dictionary that has the following keys defined:
  - **f\_0** : float carrier frequency in Hz
  - **chirprate** : float chirprate in Hz/s
  - nsamples : int number of fast time samples
  - npulses : int number of pulses
  - B\_IF: float bandwidth of intermediate frequency, demodulated signal.

- **pos** : float array [npulses x 3] (x,y,z) coordinates of platform, referenced to scene center, which is located at (0,0,0)
- **delta\_r**: float spacing between fast time samples. Defined here as  $\frac{c}{2B_{IF}}$  where,  $B_{IF}$  is the bandwidth of the intermediate frequency (de-chirped) signal.  $B_{IF}$  is simply  $(t_{max} - t_{min}) \times chirprate$  where t is the demodulated fast time, as defined in subsection 2.1.
- R\_c: float array [3]
   vector to scene center at aperture center
- t : float array [nsamples] demodulated fast-time centered at t=0. The array should go from  $(-nsamples/2, nsamples/2) \times \Delta t$  where,  $\Delta t = (A/D \text{ sampling rate})^{-1}$
- af\_r : float array [npulses]column vector containing the correction for R\_c
- af\_ph: float array [npulses]column vector containing the phase correction
- **k\_r**: float array [nsamples] spatial frequencies of demodulated signal corresponding to each fast time return. For a linear FM signal, this is given by  $\frac{4\pi}{c}(f_0 + chirprate \times t)$
- (k\_y): float
   not currently available (due to circular flight path). may be added later when moComp\_to\_line is
   complete.

## 4.3 Sandia(directory)

This function reads in the Sandia \*.phs and \*.au2 files from the user supplied directoryand exports both the phs and a Python dictionary compatible with RITSAR.

#### Parameters:

• directory: string string containing directory of DIRSIG data to include the

#### Returns:

- **phase\_history** : complex array [npulses x nsamples] the complex phase history
- platform : dictionary dictionary that has the following keys defined:
  - **f\_0** : float carrier frequency in Hz
  - **chirprate** : float chirprate in Hz/s
  - nsamples : int number of fast time samples
  - npulses : intnumber of pulses
  - B\_IF: float bandwidth of intermediate frequency, demodulated signal.
  - pos: float array [npulses x 3]
     (x,y,z) coordinates of platform, referenced to scene center, which is located at (0,0,0)

- **delta\_r**: float spacing between fast time samples. Defined here as  $\frac{c}{2B_{IF}}$  where,  $B_{IF}$  is the bandwidth of the intermediate frequency (de-chirped) signal.  $B_{IF}$  is simply  $(t_{max} - t_{min}) \times chirprate$  where t is the demodulated fast time, as defined in subsection 2.1.
- R\_c: float array [3] vector to scene center at aperture center
- **t**: float array [nsamples] <u>demodulated</u> fast-time <u>centered at t=0</u>. The array should go from  $(-nsamples/2, nsamples/2) \times \Delta t$  where,  $\Delta t = (A/D \text{ sampling rate})^{-1}$
- $\mathbf{k}_{-}\mathbf{r}$ : float array [nsamples] spatial frequencies of demodulated signal corresponding to each fast time return. For a linear FM signal, this is given by  $\frac{4\pi}{c}(f_0 + chirprate \times t)$  added later when moComp\_to\_line is complete.

# References

- [1] W. G. Carrera, R. S. Goodman, and R. M. Majewski, *Spotlight Synthetic Aperture Radar*. Artech House, Inc., 1995.
- [2] D. Wahl, P. Eichel, D. Ghiglia, and J. Jakowatz, C.V., "Phase gradient autofocus-a robust tool for high resolution sar phase correction," *Aerospace and Electronic Systems, IEEE Transactions on*, vol. 30, pp. 827–835, Jul 1994.