

Signals Correlation Algorithms For Cheaper Surveys: Using Windowing Functions.

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ABSTRACT

This paper investigates the use of baseline dependent windowing functions in interferometry data to minimize the loss of signal amplitude (smearing) when the correlated data is averaged over wide bandwidth and long time. In radio interferometry smearing is reduced when a cross-correlator averages the correlated data over narrower bandwidth and shorter integration times. Unfortunately, this leads to a huge amount of data to manage and it is becoming a bottleneck for further data processing such as calibration and imaging. With future generation surveys, it is important to investigate the reduction of the output data rate. Therefore, the focus of this paper is on the use of baselines dependent windowing functions to keep smearing down at an acceptable extent and at the same time significantly suppress signals from out field of view sources, while the nominal sensitivity is conserved.

Key words: Instrumentation: interferometers, Methods: data analysis, Methods: numerical, Techniques: interferometric

1 INTRODUCTION

A radio interferometer measures complex quantities called *visibilities*, which, following the van Cittert-Zernike relation (Thompson et al. 2001), correspond to Fourier modes of the sky brightness distribution, corrupted by various instrumental and atmospheric effects. One particular effect, known as *time* and *bandwidth smearing* (or averaging) occurs when the visibilities are averaged over a time and frequency bin of non-zero extent. This unavoidably happens in the correlator (since the correlator output is, by definition, an average measurement over some interval), but also if data is further averaged post-correlation (both for purposes of compression, and to reduce computational cost).

The effect of smearing is mainly a decrease in the amplitude of off-axis sources. This is easy to understand: the visibility contribution of a point source of flux S located in the direction given by the unit vector σ is given by

$$V = S \exp \left\{ \frac{2\pi i}{\lambda} \mathbf{u} \cdot (\sigma - \sigma_0) \right\}, \quad (1)$$

where \mathbf{u} is the baseline vector, and σ_0 is the phase centre (or fringe stopping centre) of the observation. The complex phase term above rotates as a function of frequency

(due to the inverse scaling with λ) and time (due to the fact that \mathbf{u} changes with time, at least in an Earth- or orbit-based interferometer). Taking a vector average over a time/frequency bin then results in a net loss of amplitude. The effect increases with baseline length and distance from phase centre. Besides reducing apparent source flux, smearing also distorts the PSF, since different baselines (and thus different Fourier modes) are attenuated differently.

In the era of big interferometers, where computation (and thus data size) becomes one of the main cost drivers, it is in principle desirable to average the data down as much as possible, without compromising the science goals. There are natural limits to this: firstly, we still need to critically sample the uv -plane, secondly, we need to retain sufficient spectral resolution, thirdly, we don't want to average (at least pre-calibration) beyond the natural variation of the calibration parameters, and fourthly, we want to keep smearing at acceptable levels in order not to lose too much signal. In this work, we concentrate specifically on the smearing problem. Here, we can identify two regimes:

- In a compact interferometer, the maximum usable field of view (FoV) corresponds to the primary beam (PB) of the antennas; in most cases (but surveys especially) we want the effective FoV to reach this limit. This imposes an upper limit on the size of a time/frequency bin: it must be small enough to keep amplitude loss acceptably low across the entire PB FoV.

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• In VLBI, smearing is a lot more severe, so the effective FoV is determined by the smallest time/frequency bin size that a correlator can support, and is normally much smaller than the PB. Modern VLBI correlators overcome this by employing a technique called multiple phase centre correlation, where the signal is correlated relative to multiple phase centres simultaneously, thus effectively “tiling” the PB by multiple FoVs. This has a computational cost that scales linearly with the number of phase centres.

On the other hand, smearing also has a useful side effect. In interferometry, anything outside the desired FoV is unwanted signal. However, the PB pattern of any real-life antenna features sidelobes and backlobes that extend across the entire sky, albeit at a relatively faint level. The faintness makes sidelobes useless for imaging any but the brightest sources: the scientifically usable FoV is that given by the main lobe. However, the sum total signal from all the sources in the PB sidelobes, modulated by their PSF sidelobes, contributes an unwanted global background called the *far sidelobe confusion noise* (FSCN). This imposes a fundamental sensitivity limit; in older telescopes and surveys this was well below the achievable thermal noise and therefore not a worry, but modern and future observatories are capable of reaching this limit (?). Even in observations well above the FSCN, individual extremely bright radio sources such as Cyg A or Cas A can contribute confusing signal from even the most distant sidelobe: the LOFAR telescope (?) has to deal with the so-called “A-team” sources on a routine basis. Since smearing suppresses distant sources, this somewhat alleviates both the FSCN and A-team problems.

When considering a short sequence of visibilities measured by one baseline, we can think of averaging as a convolution of the true visibility by a boxcar function corresponding to the uv -extent of the averaging bin, followed by sampling at the centre of each bin. Convolution in the visibility plane corresponds to multiplication of the image by a *tapering function* that is the Fourier transform (FT) of the convolution kernel; the FT of a boxcar is a Jinc-type taper. If we consider the entire uv -plane, averaging is only a pseudo-convolution, since the different uv -bins (and thus their boxcars) will have different sizes and shapes as determined by baseline length and orientation. Still, we can qualitatively view smearing as some kind of cumulative effect of an ensemble of image-plane tapers corresponding to all the different boxcars¹.

What if we were to employ weighted averaging instead of simple averaging (whether in the correlator, or in post-processing)? This would correspond to a pseudo-convolution of the uv -plane by some ensemble of *windowing functions* (WFs), different from boxcars, which would obviously yield different image-plane tapers, and thus result in different

smearing response. Filter theory suggests that a WF can be tuned to achieve some desired tapering response. An optimal taper would be one that was maximal across the desired FoV, and minimal outside it. In this work, we apply filter theory to derive a set of correlator WFs (CWFs) that approximate this more optimal smearing behaviour. The trade-off is an increase in thermal noise, since minimum noise can only be achieved with unweighted averaging. We show that this effect can be partially mitigated through the use of *extended WFs*.

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In the era of the Square Kilometre Array (SKA) and its pathfinders, where dealing with the huge data volumes is one of the main challenges, use of CWFs potentially offers additional leverage in optimizing radio observations. Decreased smearing across the FoV allows for more aggressive data averaging, thus reducing storage and compute costs. The trade-off is a loss of sensitivity, which pushes up observational time requirements. However, sinc

N and A-team signal could, conceivably, make up for the loss in nominal sensitivity. In the VLBI regime, use of CWFs potentially offers an increase in effective FoV at a given correlator dump rate, or equivalently, the ability to tile the PB FoV with fewer phase centres, allowing for smaller correlators.

2 OVERVIEW AND PROBLEM STATEMENT

The following formalism deals with visibilities both as functions (i.e. entire distributions on the uv -plane), and single visibilities (i.e. values of those functions at a specific point). To avoid confusion between functions in functional notation and their values, we will use \mathcal{V} or $\mathcal{V}(u, v)$ to denote functions, and V to denote individual visibilities. Likewise, $\mathcal{I}(l, m)$ denotes a function on the lm -plane i.e. an image. The symbol δ always denotes a function, that is a delta-function.

Depending on whether we want to consider polarization or not, \mathcal{V} can be taken to represent either scalar (complex) visibilities, or 2×2 complex visibility matrices as per the radio interferometer measurement equation (RIME) formalism (Smirnov 2011). Likewise, \mathcal{I} can be treated as a scalar (total intensity) image, or a 2×2 brightness matrix distribution. The derivations below are valid in either case.

We shall use the symbols $\mathbf{u} = (u, v)$ or $\mathbf{u} = (u, v, w)$ to represent baseline coordinates in units of wavelength, and \mathbf{u}^m for units of metres, with $\mathbf{u} = \mathbf{u}^m/\lambda = \mathbf{u}^m\nu/c$

2.1 Visibility and relation with the sky

An interferometer array measures the quantity $\mathcal{V}(u, v, w)$, known as the visibility function. Here, the coordinates u, v and w are vector components in units of wavelength, describing the distance between two antennas p and q , called the *baseline*. The w axis is oriented towards the *phase centre* of the observation, while u points East and v North. Given a sky distribution $\mathcal{I}(l, m)$, where l, m are the direction cosines, the nominal observed visibility is given by the van Cittert-Zernike theorem (Thompson 1999) as

$$\mathcal{V}^{\text{nom}}(u, v) = \iint_{lm} \frac{\mathcal{I}(l, m)}{\sqrt{1 - l^2 - m^2}} e^{-2\pi i \phi(u, v, w)} dldm, \quad (2)$$

¹ For completeness, we should note that this “smearing taper” is not the only tapering effect at work in interferometric imaging. Firstly, antennas have a non-zero physical extent: a measured visibility is already convolved by the aperture illumination functions (AIFs) of each pair of antennas. The resulting image-plane taper is exactly what the PB is. Secondly, most imaging software employs convolutional gridding followed by an FFT, which produces an additional taper that suppresses aliasing of sources from outside the imaged region.

where $\phi(u, v, w) = ul + vm + w(n-1)$, and $n = \sqrt{1 - l^2 - m^2}$ (the $n-1$ term comes about when fringe stopping is in effect, i.e. when the correlator introduces a compensating delay to ensure $\phi = 0$ at the centre of the field, otherwise the term is simply n).

Given a pair of antennas p and q forming a baseline $\mathbf{u}_{pq} = (u_{pq}, v_{pq}, w_{pq})$, and taking into account the *primary beam* patterns $\mathcal{E}_p(l, m)$ and $\mathcal{E}_q(l, m)$ that define the directional sensitivity of the antennas, this becomes

$$\mathcal{V}_{pq}(u, v) = \iint_{lm} \frac{\mathcal{E}_p \mathcal{E}_q^H}{\sqrt{1 - l^2 - m^2}} e^{-2\pi i \phi(u, v, w)} dldm. \quad (3)$$

Assuming a small field of view ($n \rightarrow 1$) and/or a coplanar array ($w = 0$), this becomes a 2D Fourier transform (FT):

$$\mathcal{V}_{pq}(u, v) = \iint_{lm} \mathcal{E}_p \mathcal{E}_q^H e^{-2\pi i (ul + vm)} dldm. \quad (4)$$

The effect of the primary beam can alternatively be expressed in terms of a convolution with its FT, the *aperture illumination function* (AIF) $\mathcal{A}_p(u, v)$. In functional form:

$$\mathcal{V}_{pq} = \mathcal{A}_p \circ \mathcal{V}_{pq}^{\text{nom}} \circ \mathcal{A}_q^H. \quad (5)$$

2.2 Imaging, averaging and convolution

Earth rotation causes the baseline to rotate in time, which we can denote by $\mathbf{u}_{pq}^m = \mathbf{u}_{pq}^m(t)$. The baseline in units of wavelength can be treated as a function of frequency and time:

$$\mathbf{u}_{pq}(t, \nu) = \mathbf{u}_{pq}^m(t) \nu / c. \quad (6)$$

This, in turn, allows us to express the visibility in eq. (4) as a continuous function of t, ν :

$$\mathcal{V}_{pq}(t, \nu) = \iint_{lm} \mathcal{E}_p \mathcal{E}_q^H e^{-2\pi i (u_{pq}(t)l + v_{pq}(t)m)} dldm. \quad (7)$$

Synthesis imaging recovers a so-called “dirty image” as the inverse Fourier transform of some measured visibility distribution $\mathcal{V}^{(m)}$. This is sampled by a number of baselines pq and time/frequency points t_k, ν_l . Designating each such sample by the index $pqkl$, we can express the imaging process as

$$\mathcal{I} = \mathcal{F}^{-1}[\mathcal{W} \cdot \mathcal{V}^{(m)}] \quad (8)$$

where \mathcal{W} is the (weighted) sampling function – a “bed-of-nails” function that is non-zero at points where we are sampling a visibility, and zero elsewhere. Since the Fourier transform is linear, this can be represented by a sum of “single-nail” functions \mathcal{W}_{pqkl} :

$$\mathcal{W} = \sum_{pqkl} \mathcal{W}_{pqkl} = \sum_{pqkl} W_{pqkl} \delta_{pqkl}, \quad (9)$$

where δ_{pqkl} is a delta-function shifted to the uv -point being sampled:

$$\delta_{pqkl}(u, v) = \delta(u - u_{pq}(t_k, \nu_l), v - v_{pq}(t_k, \nu_l)) \quad (10)$$

and W_{pqkl} is the associated weight. The Fourier transform being linear, we can rewrite eq. (8) as

$$\mathcal{I} = \sum_{pqkl} W_{pqkl} \mathcal{F}^{-1}[\mathcal{V}_{pqkl}^{(m)}], \quad (11)$$

where

$$\mathcal{V}_{pqkl}^{(m)} = \delta_{pqkl} V_{pqkl}^{(m)} \quad (12)$$

i.e. is a visibility distribution corresponding to the single visibility sample $pqkl$. We can rewrite eq. (8) again as

$$\mathcal{I}^D = \sum_{pqkl} W_{pqkl} \mathcal{I}_{pqkl}^D, \quad \mathcal{I}_{pqkl}^D = \mathcal{F}^{-1}[\mathcal{V}_{pqkl}^{(m)}], \quad (13)$$

which shows that the dirty image \mathcal{I}^D is a weighted sum of dirty images corresponding to the individual visibility samples $pqkl$ (each of which is essentially a single fringe pattern).

Were we to measure instantaneous visibility samples, we would have the simple relation of

$$V_{pqkl}^{(m)} = \mathcal{V}_{pq}(t_k, \nu_l), \quad (14)$$

where the right-hand side is given by eq. (7). This results in what we'll call the *ideal* dirty image \mathcal{I}^{DI} :

$$\mathcal{I}^{DI} = \sum_{pqkl} W_{pqkl} \mathcal{I}_{pqkl}^{DI}, \quad \mathcal{I}_{pqkl}^{DI} = \mathcal{F}^{-1}[\delta_{pqkl} \mathcal{V}_{pq}], \quad (15)$$

However, an interferometer necessarily measures the average visibility over a rectangular time-frequency bin

$$\mathbf{B}_{kl} = \left[t_k - \frac{\Delta t}{2}, t_k + \frac{\Delta t}{2} \right] \times \left[\nu_l - \frac{\Delta \nu}{2}, \nu_l + \frac{\Delta \nu}{2} \right], \quad (16)$$

which, ideally, can be represented by an integration:

$$V_{pqkl}^{(m)} = \frac{1}{\Delta t \Delta \nu} \iint_{\mathbf{B}_{kl}} \mathcal{V}_{pq}(t, \nu) d\nu dt. \quad (17)$$

Inverting the relation of eq. (6), we can change variables to express this as an integration over the corresponding bin \mathbf{B}'_{kl} in uv -space:

$$V_{pqkl}^{(m)} = \frac{1}{\Delta t \Delta \nu} \iint_{\mathbf{B}'_{kl}} \mathcal{V}_{pq}(u, v) \left| \frac{\partial(t, \nu)}{\partial(u, v)} \right| du dv, \quad (18)$$

where \mathbf{B}'_{kl} is the corresponding bin in uv -space. While \mathbf{B}_{kl} is rectangular in $t\nu$ space, \mathbf{B}'_{kl} is an arc segment in uv -space. Assuming a bin small enough that $\partial \mathbf{u} / \partial t$ is approximately constant over the bin, we then have

$$V_{pqkl}^{(m)} \sim \iint_{\mathbf{B}'_{kl}} \mathcal{V}_{pq}(u, v) du dv, \quad (19)$$

Now, let us introduce a *normalized boxcar windowing function*, $\Pi_{pqkl}(t, \nu)$

$$\Pi_{pqkl}(t, \nu) = \begin{cases} \frac{1}{\Delta t \Delta \nu}, & |t| \leq \Delta t/2, \quad |\nu| \leq \Delta \nu/2 \\ 0, & \text{otherwise,} \end{cases} \quad (20)$$

using which we may re-write eq. (17) as

$$V_{pqkl}^{(m)} = \iint_{\infty} \mathcal{V}_{pq}(t, \nu) \Pi_{pqkl}(t - t_k, \nu - \nu_l) dt d\nu, \quad (21)$$

which can also be expressed as a convolution:

$$V_{pqkl}^{(m)} = [\mathcal{V}_{pq} \circ \Pi_{pqkl}](t_k, \nu_l), \quad (22)$$

Likewise, eq. (18) can also be rewritten as a convolution in uv -space:

$$V_{pqkl}^{(m)} = [\mathcal{V}_{pq} \circ \Pi'_{pqkl}](u_{pq}(t_k), v_{pq}(\nu_l)), \quad (23)$$

where Π'_{pqkl} is a boxcar-like WF that corresponds to bin

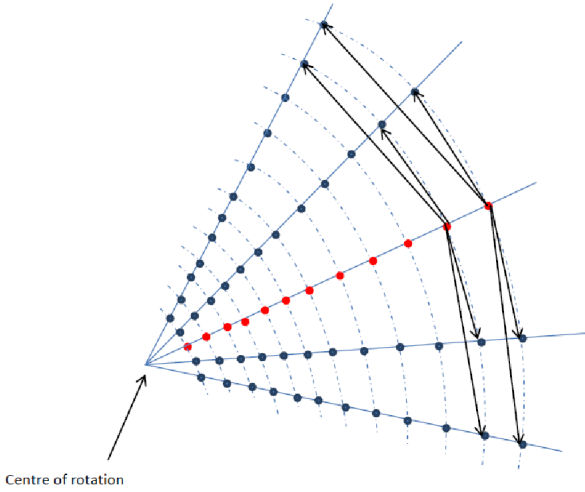


Figure 1. Schematic of uv -coverage for regularly spaced time-frequency samples.

B'_{kl} in uv -space (and also includes the determinant term of eq. 18). This makes it explicit that each averaged visibility is drawn from a convolution of the underlying visibilities with a boxcar-like WF.

Note what eq. (23) does and does not say. It does say that each individual averaged visibility corresponds to convolving the true visibilities by some WF. However, this WF is different for each baseline pq and time/frequency sample t_k, ν_l (which is emphasized by the subscripts to Π in the equations above). Averaging is thus not a “true” convolution, since the convolution kernel changes at every point in the uv -plane. We’ll call this process a *pseudo-convolution*, and the kernel being convolved with (Π'_{pqkl}) an example of a *baseline-dependent windowing function* (BDWF). In subsequent sections we will explore alternative BDWFs.

In actual fact, a correlator (or an averaging operation in post-processing) deals with averages of discrete samples, rather than a continuous integration. To represent this properly, we need to replace the integrals of eqs. (17-19) by the discrete sums:

$$V_{pqkl}^{(m)} = \sum_{i,j=-\infty}^{\infty} \mathcal{V}_{pq}(t_i, \nu_j) \Pi_{pqkl}(t_i - t_k, \nu_j - \nu_l) \quad (24)$$

and

$$V_{pqkl}^{(m)} = \sum_{i,j=-\infty}^{\infty} \mathcal{V}_{pq}(u_{ij}, v_{ij}) \Pi'_{pqkl}(u_{ij} - u_{kl}, v_{ij} - v_{kl}), \quad (25)$$

of which the first is a conventional discrete convolution (assuming a regular $t\nu$ grid), and the second is a convolution on an irregular grid – the u_{ij}, v_{ij} grid being schematically illustrated by Fig. 1.

2.3 Effect of averaging on the image

In the limit of $\Delta t, \Delta \nu \rightarrow 0$, averaging becomes equivalent to sampling. An interferometer must, intrinsically, employ a finitely small averaging interval. The Fourier phase component $2\pi\phi(u, v, w)$ is a function of frequency and time, with increasing variation over the averaging interval for sources

far from the phase centre. The average of a complex quantity with a varying phase then effectively “washes out” amplitude, the effect being especially severe for wide FoVs (for an extensive discussion, see Bregman 2012). This effect is often referred to as *time and bandwidth smearing*.

The discussion above provides an alternative way to look at smearing. Combining eqs. (8–15) with (23), and using the Fourier convolution theorem, we can see that the dirty image is formed up as

$$\mathcal{I}^D = \sum_{pqkl} W_{pqkl} \mathcal{T}_{pqkl} \mathcal{I}_{pqkl}^{DI}, \quad (26)$$

where the baseline-dependent *tapering function* \mathcal{T}_{pqkl} is the inverse FT of the BDWF:

$$\mathcal{T}_{pqkl} = \mathcal{F}^{-1}[\Pi'_{pqkl}]. \quad (27)$$

In other words, the dirty image made from averaged visibilities is a weighted average of the per-visibility ideal dirty images, each one multiplies by its own taper. The FT of a boxcar-like function is a sinc-like function, schematically illustrated in one dimension by Fig. 2. Time and bandwidth smearing represents the average effect of all these individual tapers. Shorter baselines correspond to smaller boxcars and wider tapers, longer baselines to larger boxcars and narrower tapers, and are thus more prone to smearing.

Figure 3 (produced by simulating a series of high time-frequency resolution observation using MeqTrees, and applying averaging) shows the attenuation of a 1 Jy source as a function of distance from phase centre, for a set of different time and frequency intervals. The simulations correspond to JVLA in the C configuration, with an observing frequency of 1.4 GHz. At this frequency, the first null of the PB is at $r \approx 36'$, and the half-power point is at $\sim 16'$, thus we can consider the “conventional” FoV (i.e. the half-power beam width, or HPBW) to be about 0.5° across. Note that the sensitivity of the upgraded JVLA, as well as improvements in calibration techniques, allow imaging to be done in the first PB sidelobe as well (and in fact it may be necessary for deep pointings, if only to deconvolve and subtract sidelobe sources), so we could also consider an “extended” FoV extending out to the second null of the PB at $r \approx 1.25^\circ$. Whatever definition of the FoV we adopt, Fig. 3 shows that to keep amplitude losses across the FoV to within some acceptable threshold, say 1%, the averaging interval cannot exceed some critical size, say 10s and 1 MHz. Conversely, if we were to adopt an aggressive averaging strategy for the purposes of data compression, say 50s and 5 MHz, the curves indicate that we would suffer substantial amplitude loss towards the edge of the FoV.

Finally, note that the curves corresponding to acceptably low values of smearing across the FoV (i.e. up to 25s and up to 1.25 MHz) have a very gentle slope, with very little suppression of sources *outside* the FoV.

2.4 The case for alternative BDWFs

The tapering response induced by normal averaging (Fig. 3) is far from ideal: it either suppresses too much within the FoV, or too little outside the FoV, or both. The optimal tapering response would be boxcar-like, as in Fig. 2(left). The BDWF that would produce such a response is sinc-like, as in Fig. 2(right). The problem with a sinc is that it has infinite

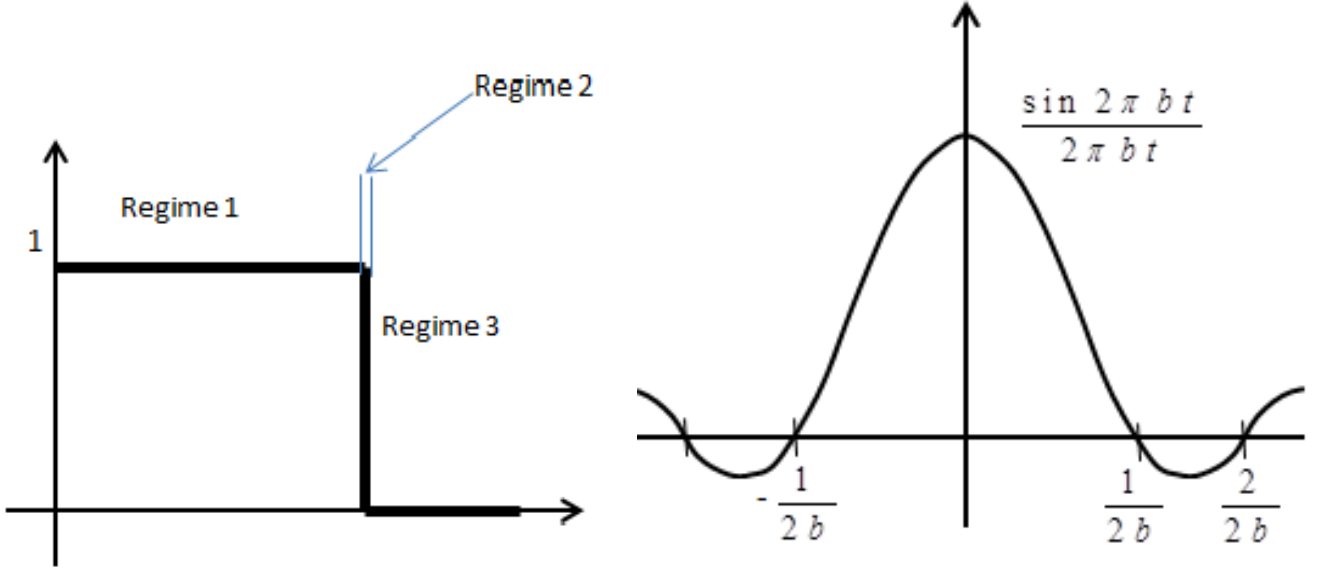


Figure 2. Left: boxcar response. In the uv -plane, this represents the windowing function corresponding to normal averaging of visibilities. In the image plane, this represents the ideal image-plane tapering function. Right: Sinc response. In the image plane, this represents the tapering function corresponding to a boxcar WF in the uv -plane. In the uv -plane, this represents the ideal WF.

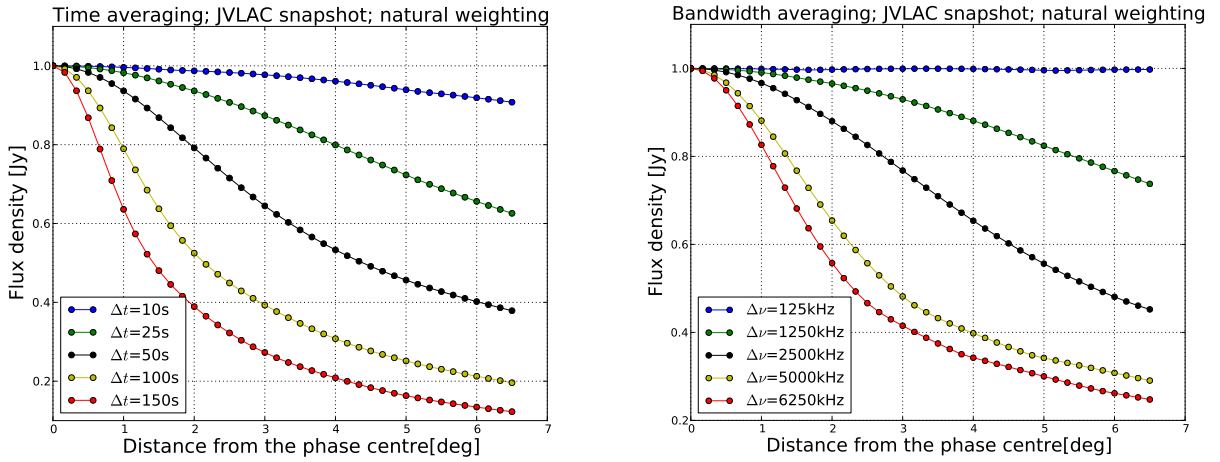


Figure 3. Effects of time and frequency averaging: the apparent intensity of a 1 Jy source, as seen by JVLAC at 1.4 GHz, as a function of distance from phase centre. (Left) Frequency interval fixed at 125 kHz, time interval varies; (right) time interval fixed at 1s, frequency interval varies.

support; applying it over finite-sized bins necessarily means a *truncated* BDWF that results in a suboptimal taper. The problem of optimal filtering has been well studied in signal processing (usually assuming a true convolution rather than the pseudo-convolution we deal with here), and we shall apply these lessons below.

The derivations above make it clear that by using a different BDWF in place of the conventional boxcar-like Π' could in principle yield more optimal tapering response. The obvious catch is a loss in sensitivity. Each visibility sample is subject to an independent Gaussian noise term in the real and imaginary part; the noise of the average of a set of samples is minimized when the average is naturally weighted (or

unweighted, if the noise is constant across visibilities). Thus, any deviation from a boxcar WF must necessarily increase the noise in the visibilities. Below we will study this effect both theoretically and via simulations, to establish whether this trade-off is sensible, and under which conditions.

3 OVERVIEW OF WINDOWING FUNCTIONS

In signal processing, a WF is a mathematical function with limited support (i.e. zero outside some interval). Conventionally, a time series is convolved with a WF to produce some desired response in the frequency domain. Applying

Signal processing	BDWFs
Frequency (freq) domain	Image plane
Time domain	Fourier plane
Spectral response	
or freq response	Image plane response (IPR)
Time response	Fourier plane response
Cut-off time interval	
or time pass band	uv averaging bin
Cut-off freq interval	
or freq pass band, or main lobe	FoV
Time stop band	Outside of the uv -bin
Freq stop band	Outside of the FoV
Octave	Doubling in size

Table 1. Mapping of terminology between signal processing and BDWFs.

this to our problem can lead to quite some confusion in terminology. Table 1 provides a mapping between the terms commonly used in SP, and their conceptual equivalent in BDWFs.

WF – or rather their corresponding image-plane response (IPR) – can be characterized in terms of various metrics. Some common ones are the peak sidelobe level (PSL), the main lobe width (MLW) and the sidelobes roll-off (SLR) rate. In terms of the “ideal” IPR (Fig. 2, left), these correspond to the following desirable features:

- Maximally conserve the signal within the FoV (“regime 1” in the figure), and make the transition in “regime 2” as sharp as possible. Both of these correspond to larger MLW.
- Attenuate sources outside the FoV (“regime 3”): this corresponds to a lower PSL and higher SLR.

3.1 Boxcar window

The Boxcar window is a function that conserved a constant non-zero value inside a chosen interval and zero value outside this interval. For a cut-off time interval $[-t_a, t_a]$ the boxcar window is defined as:

$$\Pi(t) = \begin{cases} 1 & -t_a \leq t \leq t_a \\ 0 & \text{otherwise} \end{cases} \quad (28)$$

Figures 5 and 6 give the graph of $\Pi(t)$ and its IPR respectively. The blue and the red curves of Figure 6 are the IPRs of $\Pi(t)$ for a cut-off time interval $[-t_a, t_a]$ and $[-t_a/2, t_a/2]$ respectively. Note that when the cut-off time interval is large, the MLW of the IPR is narrower, the PSL is lower and the SLR drops faster.

3.2 Gaussian window

A Gaussian window centered at mean zero with standard deviation σ is given by:

$$G(t) = e^{-bt^2}, \quad (29)$$

where, $b = (2\sigma_1^2)^{-1}$. The Fourier transform of Eq.44 is given by $\mathcal{F}\{G(t)\} = \sqrt{\frac{b}{\pi}} e^{-c\omega^2}$, where $c = \pi^2/b$. This shows us that the Fourier transform of a Gaussian with standard deviation σ_1 is a Gaussian with a standard deviation $\sigma_2 = (2\pi\sigma_1)^{-1}$. Figure 11 and 12 give the graph of $G(t)$ and its IPR respectively, where $G(t)$ is truncated within the cut-off time

interval $[-t_a, t_a]$, with $b = 3$ for the blue curve and $b = 5$ for the red curve. Note that when the standard deviation is large, the MLW of the IPR is narrower, the PSL is higher and the SLR drops slowly compared to a smaller standard deviation.

3.3 Butterworth window

The time response of the Butterworth window is flat in the time pass band, and rolls off towards zero in the time stop band and it is characterized by two independent parameters, the cut-off time time $[-t_a, t_a]$ and the order p . The two parameters control the FoV and sidelobes attenuation. The time response of the Butterworth window is given by:

$$BW(t) = \left(1 + (t/t_a)^{2p}\right)^{-1}. \quad (30)$$

For a same time interval $[-t_a, t_a]$, three curves of $BW(t)$ ($p = 1, p = 3, p = 5$) are plotted in Figure 9 and their corresponding IPRs are plotted in Figure 10. Note that when the order p increases, the MLW of the IPR is conserved, while the PSL increase and the SLR drop faster.

3.4 Sinc Window

The sinc window is defined as follow:

$$S(t) = \text{sinc}(\pi bt). \quad (31)$$

Figures 7 and 8 give the graph of $S(t)$ and its IPR respectively, where $S(t)$ is truncated within the time interval $[-t_a, t_a]$. Note that when the time interval is large (see Figure 8, blue curve), the IPR becomes perfectly flat at the frequency pass band (FoV) while the MLW becomes narrower, the PSL becomes lower and the SLR drops faster compared to a narrower time interval $[-t_a/2, t_a/2]$ (see Figure 8, red curve).

3.5 Bessel Function of the First Kind of order zero

The function can be determined using the infinity power series expansion:

$$J_0(t) = \sum_{k=0}^{\infty} \frac{(-1)^k (x/2)^{2k}}{(k!)^2} \quad (32)$$

Figures 13 and 14 give the graph of $J_0(t)$ and its IPR respectively, where $J_0(t)$ is truncated within the time interval $[-t_a, t_a]$. Note that when the time interval is large (Figure 14, blue curve), the IPR is flat at the frequency pass band (FoV) while the MLW becomes narrower, the PSL becomes lower and the SLR drops faster compared to a narrower time interval $[-t_a/2, t_a/2]$ (Figure 14, red curve).

3.6 Convolution theorem and WFs comparison

We begin with a continuous signal V_t (supposed the data is only time dependent) and sample it in order to obtain the set of measurement (see section 2.2). When the signal is limited in the time interval $[-t_a, t_a]$ and convolved with a window W_t , it follows from the convolution theorem that the process was equivalent to “tapering the Fourier transform of

the signal" by effectively looking at the true Fourier transform of the signal through the convolution window. This is presented mathematically as:

$$\mathcal{F}\{V_t \circ W_t\} = \mathcal{F}\{V_t\} \cdot \mathcal{F}\{W_t\} \quad (33)$$

One can replaced the convolution window, W_t by its Fourier transform. A summary of the MLW, the PSL and the SLR of the windows study is showed in table below.

$ \mathcal{F}\{W_t\} $	MLL (-3db)	PSL (db)	SLR (db/octave)
$\Pi(t)$	$\approx 0,073$	-6,68	-6,78
$S(t)$	$\approx 0,306$	-11,22	-12,42
$G(t)$	$\approx 0,0736$	-30,28	-14,5
$BW(t)$	$\approx 0,079$	-10,08	-15,39
$J_0(t)$	$\approx 0,39$	-15,01	-14,01

The sinc, the Bessel of the first kind of order zero (J_0) and the Butterworth (BW) approach the specification of this research and they are taken under consideration in the rest of this paper. We considered $\text{sinc}(X,Y)=\text{sinc}(X)\text{sinc}(Y)$, $J_0(X,Y)=J_0(\sqrt{X^2+Y^2})$ and $\text{BW}(X,Y)=\text{BW}(\sqrt{X^2+Y^2})$ for two dimensional sinc, J_0 and BW respectively.

4 IMAGING AND NOISE ESTIMATE

Recall from the previous section that, the boxcar WF can be replaced by a BDWF, $W_{pq,(t,\nu)}$ that depends on (u,v) distances. Now, consider that $\mathbf{W}_{pq,(t,\nu)}$ is $n_t \times n_\nu$ matrix of elements $W_{pq,(t_i,\nu_j)} \Big|_{i=1,n_t}^{j=1,n_\nu}$, the weights of uv-bins. A uv-bin is a set of four Stocks elements (XX, XY YX, YY) each with the same weight $W_{pq,(t_i,\nu_j)}$. Therefore, the sampled visibilities can be presented mathematically as a $4 \times n_t \times n_\nu$ matrix of four polarization times and frequencies dependent matrices each of size $n_t \times n_\nu$.

$$\mathbf{V}_{pq,(t,\nu)}^{smp,i} = \left(\mathbf{V}_{pq,(t,\nu)}^0, \mathbf{V}_{pq,(t,\nu)}^1, \mathbf{V}_{pq,(t,\nu)}^2, \mathbf{V}_{pq,(t,\nu)}^3 \right)^\dagger,$$

where the symbol † stand for the transpose operation. The convolution operator is linear, therefore we can re-write Eq.?? in terms of a series of linear transformations or functional models as:

$$\mathbf{V}_{pq,(t_c,\nu_c)}^{corr} = \mathbf{C}_{pq,(t,\nu)}^{block} \cdot \mathbf{W}_{pq,(t,\nu)}^{block} \cdot \mathbf{V}_{pq,(t,\nu)}^{smp,i}. \quad (34)$$

Here, $\mathbf{C}_{pq,(t,\nu)}^{block}$ and $\mathbf{W}_{pq,(t,\nu)}^{block}$ are blocks diagonals matrices of size $(4n_t n_\nu) \times (4n_t n_\nu)$, the block elements are $\mathbf{W}_{pq,(t,\nu)}$ and $\mathbf{C}_{pq,(t,\nu)}$ respectively, where $\mathbf{C}_{pq,(t,\nu)}$ is the centre time and frequency interval sampling matrix of size $n_t \times n_\nu$. This is the result of one time and frequency integration. For a synthesis, the baseline (p,q) made a complete coverage in the (u,v) plane. Therefore, we can package into a single matrix, $\mathbf{V}_{pq,(t',\nu')}^{corr}$ of size $(4N_t N_\nu) \times (4N_t N_\nu)$ the BDWF average visibilities of the baseline (p,q) during the synthesis as follows:

$$\mathbf{V}_{pq,(t',\nu')}^{corr} = \mathbf{C}_{pq,(t,\nu)}^{block,nblock} \cdot \mathbf{W}_{pq,(t,\nu)}^{block,nblock} \cdot \mathbf{V}_{pq,(t,\nu)}^{smp,nblock}. \quad (35)$$

The size of $\mathbf{V}_{pq,(t',\nu')}^{corr}$ can also be written as $(4N_v^{pq}) \times (4N_v^{pq})$, where N_v^{pq} is the total number of time and frequency visibilities (bins) for the baseline (p,q). The matrices $\mathbf{C}_{pq,(t,\nu)}^{block,nblock}$

and $\mathbf{W}_{pq,(t,\nu)}^{block,nblock}$ are diagonals blocks matrices of size $(4N_v^{pq} n_t n_\nu) \times (4N_v^{pq} n_t n_\nu)$ where each diagonal block is the block diagonal matrix $\mathbf{C}_{pq,(t,\nu)}^{block}$ and $\mathbf{W}_{pq,(t,\nu)}^{block}$ respectively. The number of block elements is n_{block} . The sampled visibilities $\mathbf{V}_{pq,(t,\nu)}^{smp,nblock} = \mathbf{S}_{pq,(t,\nu)}^{nblock} \cdot \mathbf{V}_{pq,(t,\nu)}^{smp,i}$ is a one row matrix of size $(N_v^{pq} 4n_t n_\nu) \times (4n_t n_\nu)$ made of $\mathbf{V}_{pq,(t,\nu)}^{smp,i}$ on top of each other and the matrix $\mathbf{S}_{pq,(t,\nu)}^{nblock}$ is the sampling function for the visibilities $\mathbf{V}_{pq,(t,\nu)}^{nblock}$ of size $(N_v^{pq} 4n_t n_\nu) \times (4n_t n_\nu)$. Note that $i = 1, \dots, N_v^{pq}$ and we can write:

$$\mathbf{V}_{pq,(t',\nu')}^{corr} = \mathbf{C}_{pq,(t,\nu)}^{block,nblock} \cdot \mathbf{W}_{pq,(t,\nu)}^{block,nblock} \cdot \mathbf{S}_{pq,(t,\nu)}^{nblock} \cdot \mathbf{F} \cdot \mathcal{I}_{l,m}^{sky} + \epsilon_{pq}, \quad (36)$$

our data is always corrupted by a random error component or noise, ϵ_{pq} (the error component of the baseline (p,q)). If the number of pixel in the sky model is N_{pix} , then the true sky image vector $\mathcal{I}_{l,m}^{sky}$ has a size of $4N_{pix}$ and \mathbf{F} is a block diagonal Fourier transform operator of size $(4N_{pix}) \times (4N_{pix})$.

We are generally interested in using the total set of visibilities over baselines, time and frequencies, having $4 \times N_v$ visibilities measured over all baselines with $N_v = N_{bl} \times N_v^{pq}$ where N_{bl} is the number of baseline. We then have:

$$\mathbf{V}_{array,(t',\nu')}^{corr} = \mathbf{A} \cdot \mathcal{I}_{l,m}^{sky} + \epsilon. \quad (37)$$

Here, ϵ is the array error component. \mathbf{A} is a *design* matrix of size $(4N_v) \times (4N_{pix})$ designed from $\mathbf{C}_{pq,(t,\nu)}^{block,nblock} \cdot \mathbf{W}_{pq,(t,\nu)}^{block,nblock} \cdot \mathbf{S}_{pq,(t,\nu)}^{nblock} \cdot \mathbf{F} \Big|_{p=0,\dots,n_a-1}^{q=p+1,\dots,n_a-1}$ on top of each order, with n_a the number of antenna. The *design* matrix is defined as:

$$\mathbf{A} = \begin{bmatrix} \mathbf{C}_{01,(t,\nu)}^{block,nblock} \cdot \mathbf{W}_{01,(t,\nu)}^{block,nblock} \cdot \mathbf{S}_{01,(t,\nu)}^{nblock} \cdot \mathbf{F} \\ \vdots \\ \mathbf{C}_{ik,(t,\nu)}^{block,nblock} \cdot \mathbf{W}_{ik,(t,\nu)}^{block,nblock} \cdot \mathbf{S}_{ik,(t,\nu)}^{nblock} \cdot \mathbf{F} \\ \vdots \\ \mathbf{C}_{jl,(t,\nu)}^{block,nblock} \cdot \mathbf{W}_{jl,(t,\nu)}^{block,nblock} \cdot \mathbf{S}_{jl,(t,\nu)}^{nblock} \cdot \mathbf{F} \end{bmatrix}$$

The dirty image, $\mathcal{I}_{l,m}^D$ of size $4N_{pix}$ can then be derived as follow:

$$\mathcal{I}_{l,m}^D = \left(\mathbf{F}^H \cdot \mathbf{A} \cdot \mathcal{I}_{l,m}^{sky} \right)_{(l,m)} + \epsilon. \quad (38)$$

Here, H represents the the conjugate transpose operation also known as a Hermitian transpose and \mathbf{F}^H is a block diagonal inverse Fourier transform operator of size $(4N_{pix}) \times (4N_{pix})$.

The estimate of ϵ , for the map centre pixel is given by:

$$\tilde{\epsilon}_{o,o} = \tilde{\mathcal{I}}_{o,o} - \left(\mathbf{F}^H \cdot \mathbf{A} \cdot \tilde{\mathcal{I}} \right)_{(o,o)} \quad (39)$$

$$= \frac{1}{N_v} \left(1 - \left(\mathbf{F}^H \cdot \mathbf{A} \right)_{(o,o)} \right) \left(\mathbf{B}^\dagger \cdot \mathbf{V}_{array,(t,\nu)}^{smp} \right). \quad (40)$$

Here, \mathbf{B} and $\mathbf{V}_{array,(t,\nu)}^{smp}$ are one row matrix of size $(N_v 4n_t n_\nu) \times (4n_t n_\nu)$ made of $\mathbf{C}_{pq,(t,\nu)}^{block,nblock}$.

$\mathbf{W}_{pq,(t,\nu)}^{block,nblock} \Big|_{p=0,\dots,n_a-1}^{q=p+1,\dots,n_a-1}$ on top of each other and $\mathbf{V}_{pq,(t,\nu)}^{smp,nblock} \Big|_{p=0,\dots,n_a-1}^{q=p+1,\dots,n_a-1}$ on top of each other respectively (see appendix A for the derivation of these complex matrices). For a noisy sky model, $\mathbf{V}_{array,(t,\nu)}^{smp} = \sigma \mathbf{G}$, where σ is

the expected rms noise per visibility before boxcar averaging or BDWF averaging and \mathbf{G} is a $(N_v 4n_t n_\nu) \times (4n_t n_\nu)$ unit matrix. Eq.40 can therefore be rewritten as:

$$\tilde{\epsilon}_{o,o} = \frac{\sigma}{N_v} \left(1 - \left(\mathbf{F}^H \cdot \mathbf{A} \right)_{(o,o)} \right) \left(\mathbf{B}^\dagger \cdot \mathbf{G} \right). \quad (41)$$

The formalism of Eq.41 will be used to derive the noise estimate in section 6.7.

5 FOV SIGNALS RECOVERY AND OUT FIELD SUPPRESSION

5.1 FoV signals recovery: description and method

Missing spaces between sampled (u, v) coordinates have an important dependence on the baseline length. However, the spacings between longer baselines (u, v) coordinates are wider than the one on shorter baselines, this is the obvious effect that explained why sources are more distorted on longer baselines compared to shorter ones. Therefore, if one has to attribute the weight of a uv-bin, it may be proportional to the baseline length, in such a way that the distortion rate is taken into account over baselines. The study in this section aims to describe an approach that used a WF to assign a proper weight of a uv-bin considering the *spacing* between the baseline uv-bins and the centre uv-bin.

Figure 1 shows a snapshot coverage of an integration time interval. For shorter baselines, the tracks are closer to the centre of rotation and for longer baselines the tracks are farther away from the centre. The *dot marks* are the uv-bins, and the arrows indicate the separation between uv-bins and the centre uv-bin. It is trivial to see on this figure that these separations are wider on longer baselines. The results of boxcar averaging or BDWF averaging are assigned to the centre uv-bin coloured in red.

During integration, the Earth rotation makes baselines coordinates to vary in time and frequency. We can therefore package the (u, v) coordinates changes and the frequency changes of a baseline (p,q) into a matrix of size $n_t \times 2$ and vector of dimension n_ν respectively.

$$\mathbf{U}_{pq,t} = \begin{pmatrix} \mathbf{u}_{pq,t_s}, \dots, \mathbf{u}_{pq,t_c}, \dots, \mathbf{u}_{pq,t_e} \end{pmatrix}^\dagger$$

$$\boldsymbol{\nu} = \begin{pmatrix} \nu_s, \dots, \nu_c, \dots, \nu_e \end{pmatrix}^\dagger.$$

The elements of $\mathbf{U}_{pq,t}$ are functions of time and frequency representing the (u, v) coordinates. A function $\bar{\cdot}$ on a $n_t \times 2$ matrix is defined as follow:

$$\bar{\bar{\mathbf{U}}}_{pq,t} = \left(\|\mathbf{u}_{pq,t_s}\|, \dots, \|\mathbf{u}_{pq,t_c}\|, \dots, \|\mathbf{u}_{pq,t_e}\| \right)^\dagger, \quad (42)$$

where $\|\cdot\|$ is the Euclidean norm.

Definition 1.1. (Time direction spacing) The vector of size n_t that models the spacing between uv-bins and the centre uv-bin of a baseline (p,q) across the time direction is defined as:

$$\mathbf{d}_t = \frac{\nu_c}{c} \cdot \left\{ \bar{\bar{\mathbf{U}}}_{pq,t} - \mathbf{H}_{pq,t} \right\},$$

where c is the speed of the light in m/s and $\mathbf{H}_{pq,t}$ is a matrix of size $n_t \times 2$ that models the centre uv-bin,

$$\mathbf{H}_{pq,t} = (\mathbf{u}_{pq,t_c}, \dots, \mathbf{u}_{pq,t_c}, \dots, \mathbf{u}_{pq,t_c})^\dagger.$$

Definition 1.2. (Frequency direction spacing) The vector of size n_ν that models the spacing between uv-bins and the centre uv-bin of a baseline (p,q) across the frequency direction is defined as:

$$\mathbf{d}_\nu = \frac{\|\mathbf{u}_{pq,t_c}\|}{c} \cdot \left\{ \boldsymbol{\nu} - \nu_c \cdot \mathbf{g}_\nu \right\},$$

where \mathbf{g}_ν is a $n_\nu \times 1$ unit matrix.

Definition 1.3. (Baseline dependent windowing function)

If f_{pq} is a *baseline dependent windowing function*, then:

$$f_{pq} : \{\mathcal{R}, \mathcal{R}\} \rightarrow \mathcal{R}$$

$$d_{t_i}, d_{\nu_j} \mapsto \frac{w_{t_i, \nu_j}}{\sum_{i=1}^{n_t} \sum_{j=1}^{n_\nu} w_{t_i, \nu_j}}.$$

Here d_{t_i} is an element of the vector \mathbf{d}_t and d_{ν_j} is an element of the vector \mathbf{d}_ν .

The previous approach is exact but unfortunately, we do not suppress confusion sources and we lost in sensitivity. Although boxcar averaging *means high sensitivity*, we do need to attenuate confusion sources in such a way that the overall Signal to Noise Ratio (SNR) becomes higher than that of boxcar averaging. The following section describe such method.

5.2 Out FoV suppression: description and method

In theory, WFs and signals generally extend to infinity. Unfortunately, in practice, filtering a signal with a low pass filter, one need to define a cut-off interval (range of uv-bins). Therefore, if one wants to achieve sufficiently an accurate estimate of the WF ideal image plane response (IPR), one need a wide cut-off interval as far as the IPR approaches the ideal when the WF order increases. An overlap BDWF aims to extend the order of the BDWF in such a way that, we approach the ideal IPR. The only drawback of this technique is the increase in time needed for processing the output sample of the signals being integrate.

The weight of a uv-bin is not defined by a unique BDWF but by the strength of the correlation between the overall overlapping BDWF as shown in Figure 4. That said, the weight of a uv-bin becomes the summation of overlapping BDWF samples on the uv-bin normalized by the summation of the overall uv-bins weights within the range where the overlap samples are defined.

6 ANALYSIS OF WINDOWING FUNCTIONS

In signal processing a WF is a mathematical function that has zero-values outside some chosen interval, and when a signal is multiplied by the WF, the product has also zero-values outside the interval. In this section, we evaluated the Peak Sidelobe Level (PSL), the Main Lobe width (MLW) and the Sidelobes Roll-off (SLR) for some WFs. Allowing this paper to be useful both in signal processing and radio astronomy community, the terminology that

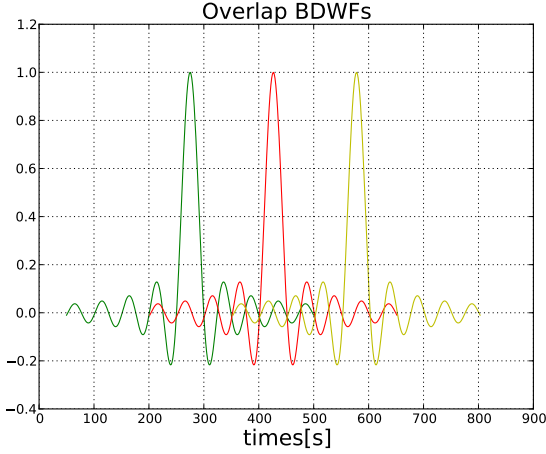


Figure 4. Overlap baseline dependent windowing functions

is used in radio astronomy in contrast of the terminology used in signal processing is presented in the table below.

Table of terms

Signal processing	BDWFs
Frequency (freq) domain	Image plane
Time domain	Fourier plane
Spectral response or freq response	Image plane response
Time response	Fourier plane response
Cut-off time interval or time pass band	uv-bins range
Cut-off freq interval or freq pass band	FoV
Time stop band	Outside uv-bins range
Freq stop band	Outside edges of FoV

The goal of this section is threefold, find a WF that its IPR will:

- Conserved the signal within a field of interest "Regime 1": that said a WF with a narrower MLW.
- Achieved less masking of nearby sources by restricting "Regime 2" to a narrower band: that said a WF with a lower PSL.
- Attenuated sidelobes confusion from far away sources "Regime 3": that said a WF with a SLR that drops faster.

Such a WF is very common in theory, is a continuous function (unsampled function) over an illimitable time interval with unit area as shown in Figure 2 and its ideal IPR in Figure ??.

6.1 Boxcar window

The Boxcar window is a function that conserved a constant non-zero value inside a chosen interval and zero value outside this interval. For a cut-off time interval $[-t_a, t_a]$ the boxcar window is defined as:

$$\Pi(t) = \begin{cases} 1 & -t_a \leq t \leq t_a \\ 0 & \text{otherwise} \end{cases} \quad (43)$$

Figures 5 and 6 give the graph of $\Pi(t)$ and its IPR respectively. The blue and the red curves of Figure 6 are the IPRs of $\Pi(t)$ for a cut-off time interval $[-t_a, t_a]$ and $[-t_a/2, t_a/2]$

respectively. Note that when the cut-off time interval is large, the MLW of the IPR is narrower, the PSL is lower and the SLR drops faster.

6.2 Gaussian window

A Gaussian window centered at mean zero with standard deviation σ is given by:

$$G(t) = e^{-bt^2}, \quad (44)$$

where, $b = (2\sigma_1^2)^{-1}$. The Fourier transform of Eq.44 is given by $\mathcal{F}\{G(t)\} = \sqrt{\frac{b}{\pi}} e^{-cl^2}$, where $c = \pi^2/b$. This shows us that the Fourier transform of a Gaussian with standard deviation σ_1 is a Gaussian with a standard deviation $\sigma_2 = (2\pi\sigma_1)^{-1}$. Figure 11 and 12 give the graph of $G(t)$ and its IPR respectively, where $G(t)$ is truncated within the cut-off time interval $[-t_a, t_a]$, with $b = 3$ for the blue curve and $b = 5$ for the red curve. Note that when the standard deviation is large, the MLW of the IPR is narrower, the PSL is higher and the SLR drops slowly compared to a smaller standard deviation.

6.3 Butterworth window

The time response of the Butterworth window is flat in the time pass band, and rolls off towards zero in the time stop band and it is characterized by two independent parameters, the cut-off time time $[-t_a, t_a]$ and the order p . The two parameters control the FoV and sidelobes attenuation. The time response of the Butterworth window is given by:

$$BW(t) = \left(1 + (t/t_a)^{2p}\right)^{-1}. \quad (45)$$

For a same time interval $[-t_a, t_a]$, three curves of $BW(t)$ ($p = 1, p = 3, p = 5$) are plotted in Figure 9 and their corresponding IPRs are plotted in Figure 10. Note that when the order p increases, the MLW of the IPR is conserved, while the PSL increase and the SLR drop faster.

6.4 Sinc Window

The sinc window is defined as follow:

$$S(t) = \text{sinc}(\pi bt). \quad (46)$$

Figures 7 and 8 give the graph of $S(t)$ and its IPR respectively, where $S(t)$ is truncated within the time interval $[-t_a, t_a]$. Note that when the time interval is large (see Figure 8, blue curve), the IPR becomes perfectly flat at the frequency pass band (FoV) while the MLW becomes narrower, the PSL becomes lower and the SLR drops faster compared to a narrower time interval $[-t_a/2, t_a/2]$ (see Figure 8, red curve).

6.5 Bessel Function of the First Kind of order zero

The function can be determined using the infinity power series expansion:

$$J_0(t) = \sum_{k=0}^{\infty} \frac{(-1)^k (x/2)^{2k}}{(k!)^2} \quad (47)$$

Figures 13 and 14 give the graph of $J_0(t)$ and its IPR respectively, where $J_0(t)$ is truncated within the time interval $[-t_a, t_a]$. Note that when the time interval is large (Figure 14, blue curve), the IPR is flat at the frequency pass band (FoV) while the MLW becomes narrower, the PSL becomes lower and the SLR drops faster compared to a narrower time interval $[-t_a/2, t_a/2]$ (Figure 14, red curve).

6.6 Convolution theorem and WFs comparison

We begin with a continuous signal V_t (supposed the data is only time dependent) and sample it in order to obtain the set of measurement (see section 2.2). When the signal is limited in the time interval $[-t_a, t_a]$ and convolved with a window W_t , it follows from the convolution theorem that the process was equivalent to "*tapering the Fourier transform of the signal*" by effectively looking at the true Fourier transform of the signal through the convolution window. This is presented mathematically as:

$$\mathcal{F}\{V_t \circ W_t\} = \mathcal{F}\{V_t\} \cdot \mathcal{F}\{W_t\} \quad (48)$$

One can replaced the convolution window, W_t by its Fourier transform. A summary of the MLW, the PSL and the SLR of the windows study is showed in table below.

$ \mathcal{F}\{W_t\} $	MLL (-3db)	PSL (db)	SLR (db/octave)
$\Pi(t)$	$\approx 0,073$	-6,68	-6,78
$S(t)$	$\approx 0,306$	-11,22	-12,42
$G(t)$	$\approx 0,0736$	-30,28	-14,5
$BW(t)$	$\approx 0,079$	-10,08	-15,39
$J_0(t)$	$\approx 0,39$	-15,01	-14,01

The sinc, the Bessel of the first kind of order zero (J_0) and the Butterworth (BW) approach the specification of this research and they are taken under consideration in the rest of this paper. We considered $\text{sinc}(X,Y)=\text{sinc}(X)\text{sinc}(Y)$, $J_0(X,Y)=J_0(\sqrt{X^2+Y^2})$ and $BW(X,Y)=BW(\sqrt{X^2+Y^2})$ for two dimensional sinc, J_0 and BW respectively.

The following designations will be used in the rest of the paper:

Bl-WF- $n_{ovlpt} \times -$: One dimensional BDWF applied across the time direction; WF is one of the window under consideration (sinc, J_0 and BW) and $2n_{ovlpt}$ is the number of overlap time bins.

Bl-WF- $- \times n_{ovlpv}$: One dimensional BDWF applied across the frequency direction with $2n_{ovlpv}$ the number of overlap frequency bins.

Bl-WF- $n_{ovlpt} \times n_{ovlpv}$: Two dimensional BDWF applied across both time and frequency direction.

Bx-avg- $0 \times -$: Boxcar averaging applied across the time direction.

Bx-avg- $- \times 0$: Boxcar averaging applied across the frequency direction.

Bx-avg- 0×0 : Boxcar averaging applied across both time and frequency direction.

6.7 Theoretical noise estimate and simulation noise

We show in this section that on shorter baselines, the sinc, the J_0 and the BW reduce to the Boxcar window. Therefore, we effectively applying a hybrid window in the Fourier plane; a Boxcar on the shorter baselines and the WF under consideration on the longer baselines (see Figure 15, 16 and 17). We furthermore evaluated the array theoretical noise ratio, $R_\sigma = \epsilon_{ar,oo}^{bd}/\epsilon_{ar,oo}^{bx}$ predicted in Eq.41 of a 2° BDWF averaging by the one of boxcar averaging and compared the ratio to the simulation one. For the analysis we used a JVLAC measurement set (MS) of 7min30s synthesis with a 1.5s integration time at 1.4Ghz, with 150 channels of width 125kHz. The MS is filled with 1Jy thermal noise. We then processed boxcar averaging and BDWF averaging over 1min30s (width of the time integration) and 12500kHz (channel width). Two overlaps BDWF are presented, the first overlap over 2min60s across time and 3125kHz across frequency; the second overlap over 3min45s across time and 6250kHz across frequency.

- In Figure 18, 19 and 20 each *dot mark* is the ratio ² of per baseline centre pixel theoretical rms noise ($\epsilon_{pq,oo}^{bd}$) of a 2° BDWF averaging by the one of boxcar averaging ($\epsilon_{pq,oo}^{bx}$). The ratio is plotted as a function of \bar{d}_t (the mean of the baseline (u, v) distance). The figures show that, the noise increases with baseline length.

- On shorter baselines, the noise estimate of BDWF averaging is approximately the same with the one of boxcar averaging. Also, on shorter baselines, the noise drops significantly with the number of overlap time and frequency bins. As mention above, this is because on shorter baselines, the BDWF are closer to the boxcar window, and when the number of overlap time and frequency bins increases, the ratio: $\epsilon_{pq,oo}^{bd}/\epsilon_{pq,oo}^{bx} \approx \sqrt{n_t n_\nu / ((n_t + 2n_{ovlpt})(n_\nu + 2n_{ovlpv}))}$.

The table below summarized the array theoretical rms noise ratio and the simulation one. One do not need to worry about a perfect agreement of the theoretical noise results with the simulation one. Unfortunately, the theoretical noise results are the rms noise of the sky map centre pixel while the simulation results are an average of the sky map overall pixels rms noise.

BDWF	Theoretical	Simulation
Bl-sinc- 0×0	1.17	1.23
Bl-sinc- 100×25	1.02	1.18
Bl-sinc- 150×50	1.03	1.27
Bl- J_0 - 0×0	1.11	1.14
Bl- J_0 - 100×25	0.91	1.08
Bl- J_0 - 150×50	0.92	1.13
Bl-BW- 0×0	1.16	1.55
Bl-BW- 100×25	1.44	1.49
Bl-BW- 150×50	1.42	1.44

² $\epsilon_{pq,oo}^{bd}/\epsilon_{pq,oo}^{bx}$, see Eq.36 for a recall of $\epsilon_{pq,oo}^{bd}$ and $\epsilon_{pq,oo}^{bx}$.

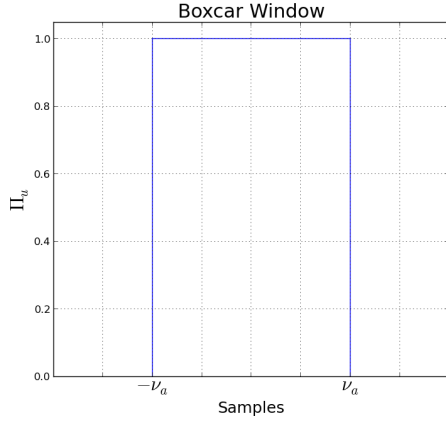


Figure 5. Boxcar windowing function.

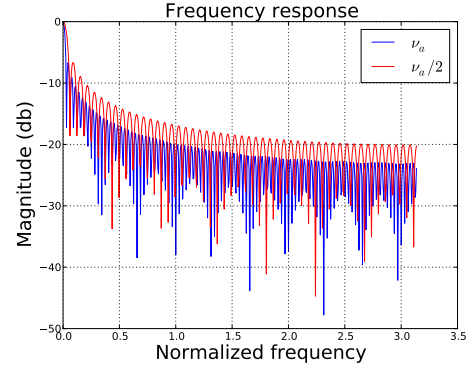


Figure 6. Frequency response of a boxcar window

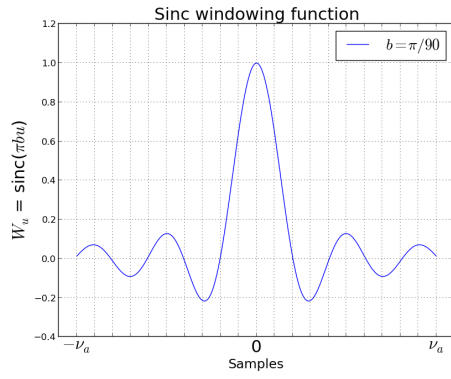


Figure 7. Sinc window

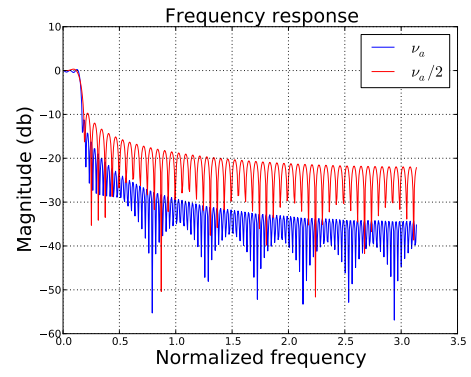


Figure 8. Frequency response of the sinc window

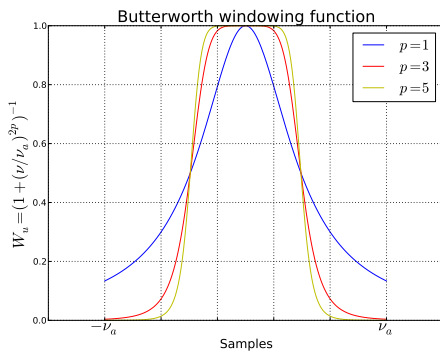


Figure 9. Butterworth windows

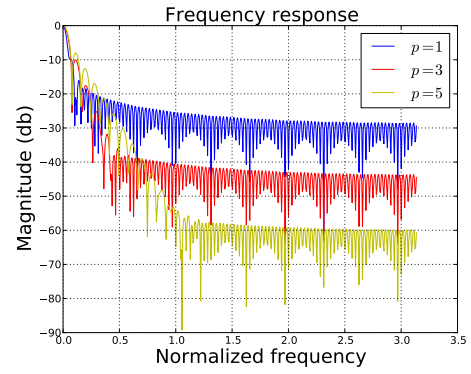


Figure 10. Frequency response of the Butterworth windows

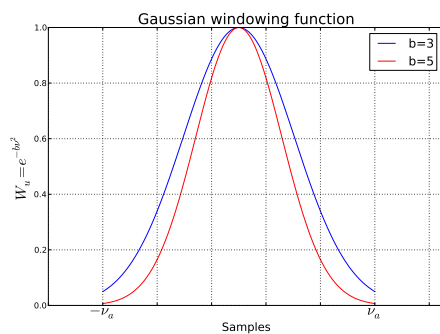


Figure 11. Gaussian windows

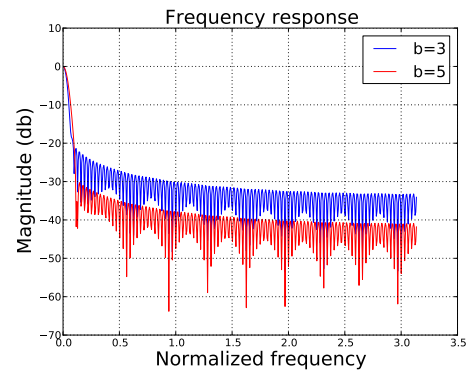


Figure 12. Frequency response of Gaussian windows

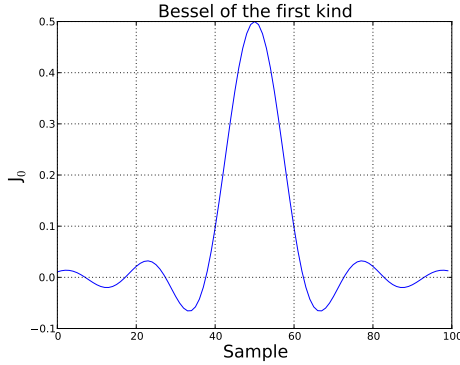


Figure 13. Bessel first King windows NB: this figure is coming very soon

7 SIMULATIONS AND RESULTS

In order to test the approaches described in sections 5.1 and 5.2, multiple tests are performed on JVLAC simulated MS. This section summarized and discussed those results. Two MS are used in the simulation, a high resolution MS (HR-MS) that contains the observed JVLAC data of short integration time and frequency and a low resolution MS (LR-MS) where the results of boxcar averaging or BDWF averaging are saved. In order to apply an overlap BDWF the following conditions have to be satisfied:

- (i) If t_{start}^{hrms} and t_{start}^{lrms} are the starting time of the HR-MS and LR-MS respectively and Δt^{hrms} the HR-MS integration time, then $t_{start}^{lrms} \geq t_{start}^{hrms} + 2n_{ovlpt} \times \Delta t^{hrms}$.
- (ii) If t_{end}^{hrms} and t_{end}^{lrms} are the ending time of the HR-MS and LR-MS respectively, then $t_{end}^{lrms} \leq t_{end}^{hrms} + 2n_{ovlpt} \times \Delta t^{hrms}$.
- (iii) If ν_{start}^{hrms} and ν_{start}^{lrms} are the starting frequency of the HR-MS and LR-MS respectively and $\Delta \nu^{hrms}$ the HR-MS channel width, then $\nu_{start}^{lrms} \geq \nu_{start}^{hrms} + 2n_{ovlpt} \times \Delta \nu^{hrms}$.
- (iv) If ν_{end}^{hrms} and ν_{end}^{lrms} are the ending frequency of the HR-MS and LR-MS respectively, then $\nu_{end}^{lrms} \leq \nu_{end}^{hrms} + n_{ovlpt} \times \Delta \nu^{hrms}$.

7.1 Smearing elimination and out FoV suppression

The study considered 40 different sky models each containing a 1Jy source; the sky models differ from each other by the source coordinates. The reason for having only one source in the sky model is to avoid standard imaging artefacts in the sky maps which can affect the brightness of sources. Another obvious reason is the CLEAN algorithm (Cornwell et al. 1999), which does not create perfect clean maps. Each sky model is simulated with a JVLAC HR-MS of 7min30s synthesis, with a $\Delta t^{hrms} = 1.5s$ integration time at 1.4GHz, with 150 channels width $\Delta \nu^{hrms} = 125kHz$. Each HR-MS is then processed with boxcar averaging and BDWF averaging the results are saved into a JVLAC LR-MS of one timeslot of with 150s, with 1 channel of width 6.25MHz. The study also considered, $\{t_{start}^{hrms}, \nu_{start}^{hrms}\} = \{0s, 125kHz\}$, $\{t_{start}^{lrms}, \nu_{start}^{lrms}\} = \{150s, 6250kHz\}$, $\{t_{end}^{hrms}, \nu_{end}^{hrms}\} = \{7min30s, 18750kHz\}$,

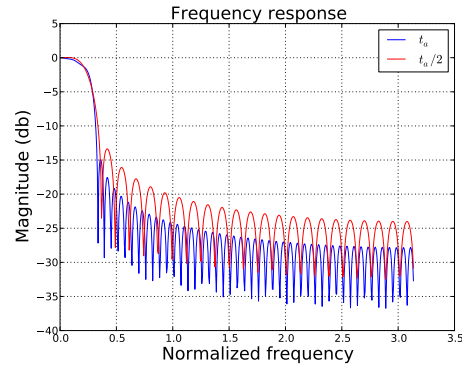


Figure 14. Frequency response of Bessel first kind window

$\{t_{end}^{lrms}, \nu_{end}^{lrms}\} = \{6min, 12500kHz\}$, $n_{ovlpt} \in \{0, 100, 150\}$ and $n_{ovlpt} \in \{0, 25, 50\}$.

The source flux density as a function of the source distance from the phase centre is depicted in Figures 21 to 26. The study in Figure 21, 23 and 25 compared boxcar averaging with 2° FoV of Bl-sinc- $n_{ovlpt} \times n_{ovlpt}$, Bl-J₀- $n_{ovlpt} \times n_{ovlpt}$ and Bl-BW- $n_{ovlpt} \times n_{ovlpt}$ respectively. A study with a 2° FoV suggest that: "Regime 1" is defined from 0° to 1° ; "Regime 2" from 1° to 1.5° and "Regime 3" from 1.5° to infinity. The study in Figure 22, 24 and 26 compared boxcar averaging with 4° FoV of Bl-sinc- $n_{ovlpt} \times n_{ovlpt}$, Bl-J₀- $n_{ovlpt} \times n_{ovlpt}$ and Bl-BW- $n_{ovlpt} \times n_{ovlpt}$ respectively. A study with a 4° FoV suggest that: "Regime 1" is defined from 0° to 2° ; "Regime 2" from 2° to 3° and "Regime 3" from 3° to infinity. The results show that:

- "Regime 1" (FoV recovery): The source flux density is recovered with the BDWF averaging compared to the boxcar averaging. Also, as the overlap time and frequency bins increase, the source flux density becomes optimal with the overlap sinc and J₀ (see the green and the red curve from Figures 21 to 24) which is not the case with the overlap BW as shown in Figures 25 and 26 (the green and the red curve).
- "Regime 2" (nearby sources suppression): The source flux density is suppressed using boxcar averaging compared to BDWF averaging without overlap bins (see Bl-sinc- 0×0 , Bl-J₀- 0×0 and Bl-BW- 0×0 shown from Figure 21 to 26). It is also seen from Figure 21 to 24 (the green and the red curve) that, the source flux density is suppressed within $\approx 25\%$ of the regime using boxcar averaging than the sinc and J₀ overlap BDWF averaging while within $\approx 75\%$ of the regime the sinc and J₀ overlap BDWF averaging suppressed the source compared to the boxcar averaging. The performance of the overlap BW BDWF averaging is not accurate in this regime.
- "Regime 3" (far away sources attenuation): It appears on Figures 21 to 24 that the curves of boxcar averaging (see Bx-avg- 0×0) mid that of Bl-sinc- 0×0 and Bl-J₀- 0×0 which implied an equivalent flux density attenuation while the curve of Bl-sinc- 100×25 , Bl-sinc- 150×50 , Bl-J₀- 100×25 and Bl-J₀- 150×50 are below the one of Bx-avg- 0×0 which implied that the overlap sinc and J₀ BDWF attenuated the source in the regime more than boxcar averaging. Also, as the overlap time and frequency bins increase, the attenuation becomes significant. The similar results occurs 1° away

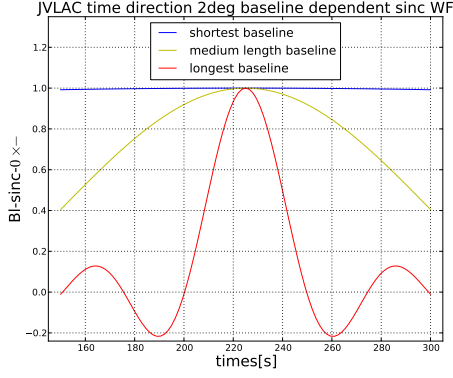


Figure 15. Time direction BI-sinc-W0 of the shortest, medium and longest baseline

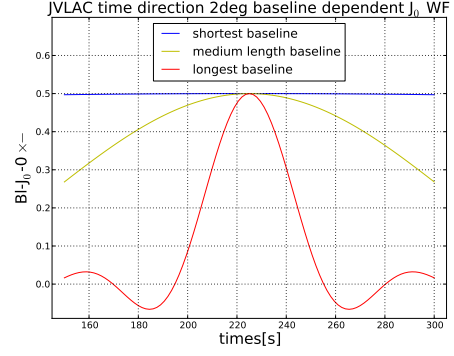


Figure 16. Time direction BI-J0-W0 of the shortest, medium and longest baseline

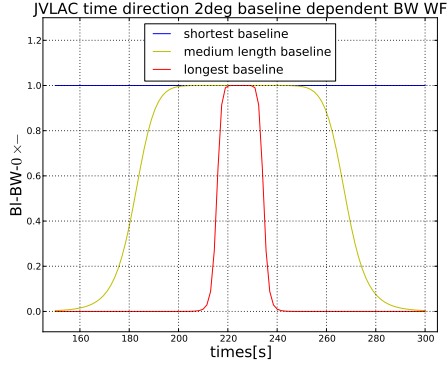


Figure 17. Time direction BI-BW-W0 of the shortest, medium and longest baseline

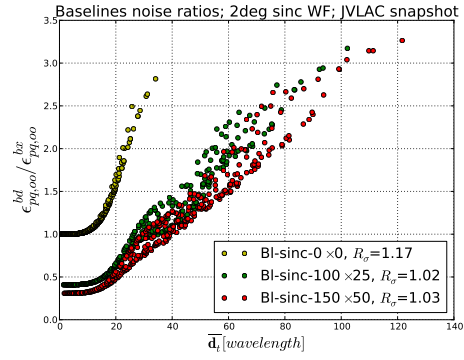


Figure 18. Per baseline noise ratio of BI-sinc- $Wn_{lt} \times n_{ltv}$ and averaging

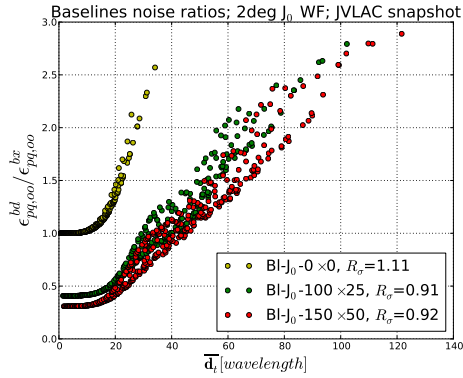


Figure 19. Per baseline noise ratio of BI-J0- $Wn_{lt} \times n_{ltv}$ and averaging

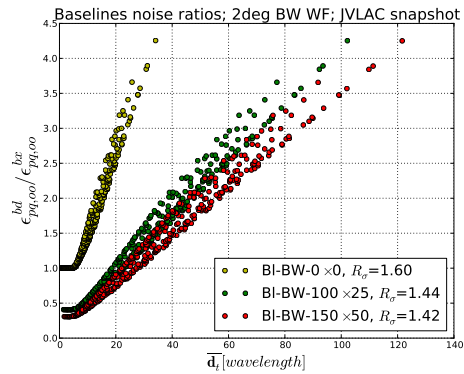


Figure 20. Per baseline noise ratio of BI-BW- $Wn_{lt} \times n_{ltv}$ and averaging

from the regime starting interval with BW BDWF averaging.

7.2 Maximum integration

In this section, the maximum frequency and time integration intervals that can be considered in the frequency and the time directions without loss of signal amplitude is eval-

uated. The study considered 6 sky models each containing 1Jy source; the sky models differ by the source coordinates with radius $r \in \{0.09^\circ, 0.25^\circ, 0.5^\circ, 1^\circ, 1.5^\circ, 2^\circ\}$. The study compared the results of boxcar averaging to a 2° BDWF averaging. As mention above, a 2° FoV suggest that: "Regime 1" is defined from 0° to 1° ; "Regime 2" from 1° to 1.5° and "Regime 3" from 1.5° to infinity. Therefore, a sky model with a source radius $r \in \{0.09^\circ, 0.25^\circ, 0.5^\circ, 1^\circ\}$ belongs to

"Regime 1", $r = 1.5^\circ$ belongs to "Regime 2" and $r = 2^\circ$ to "Regime 3".

7.2.1 Frequency direction

Boxcar and BDWF averaging are processed across the frequency direction. Each of the sky models is simulated with a JVLAC HR-MS of 1 timeslot of width $\Delta t^{hrms} = 0.1s$ at 1.4GHz. The reason of a short HR-MS integration time is to avoid time direction smearing. The simulation considered 150 channels and varied the channels width $\Delta\nu^{hrms}$ in the range [125, 1187.5]kHz. The result of the process was saved into a LR-MS of 1 timeslot of width $\Delta t^{lrms} = 0.1s$ at 1.4GHz with 1 channel of width $\Delta\nu^{hrms} \times 50kHz$. We considered $\nu_{start}^{hrms} = \Delta\nu^{hrms}kHz$, $\nu_{start}^{lrms} = \nu_{start}^{hrms} \times 50kHz$, $\nu_{end}^{hrms} = \nu_{start}^{hrms} \times 150kHz$ and $\nu_{end}^{lrms} = \nu_{start}^{hrms} \times 100kHz$.

The source flux density as a function of the channel width is depicted in Figures 28 to 33. The study compared the maximal channel width of the boxcar frequency averaging (Bx-avg- $\times 0$) with that of 2° FoV BDWF frequency averaging (Bl-sinc- $\times 0$, Bl-sinc- $\times 50$, Bl-J₀- $\times 0$, Bl-J₀- $\times 50$, Bl-BW- $\times 0$ and Bl-BW- $\times 50$). The results show that:

- "Regime 1": The BDWF frequency averaging maintained the flux density of the source for wider frequency channels. However, it is seen on the Figures that the curve of sources in the regime are closer to the affine line $y=1$ when the channel width increases.
- "Regime 2" and "Regime 3": It appears on the Figures that Boxcar frequency averaging attenuated the sources in these regime compare to BDWF frequency averaging without overlap while overlap sinc and J₀ frequency averaging attenuated the source for channels widths between [6.2, 12.6] compare to the boxcar.

7.2.2 Time direction

Boxcar and BDWF averaging are processed across the time direction. Each of the sky models is simulated with a JVLAC HR-MS of 300 timeslots of width $\Delta t^{hrms}s$ at 1.4GHz. The value of Δt^{hrms} is chosen within the range [0.5, 5.5]. The simulation considered 1 channel of width $\Delta\nu^{hrms} = 125kHz$ and the results of the process was saved into a LR-MS of $100 \times \Delta t^{hrms}s$ synthesis, with a $100 \times \Delta t^{hrms}s$ integration time at 1.4GHz, with 1 channel of width $125kHz$. We considered $t_{start}^{hrms} = 0s$, $t_{start}^{lrms} = 100 \times \Delta t^{hrms}s$, $t_{end}^{hrms} = 300 \times \Delta t^{hrms}s$ and $t_{end}^{lrms} = 200 \times \Delta t^{hrms}s$.

The source flux density as a function of the integration time width is depicted in Figures 34 to 39. The study compared the maximal integration time width of the boxcar time averaging (Bx-avg- $0 \times -$) with that of 2° FoV BDWF time averaging (Bl-sinc- $100 \times -$, Bl-sinc- $100 \times -$, Bl-J₀- $0 \times -$, Bl-J₀- $100 \times -$, Bl-BW- $0 \times -$ and Bl-BW- $100 \times -$). The results show that:

- "Regime 1": The BDWF time averaging maintained the flux density of the source for large integration time. However, it is seen on the Figures that the curve of sources in the regime are closer to the affine line $y=1$ when the integration time increases.
- "Regime 2" and "Regime 3": It appears on the Figures

that Boxcar time averaging attenuated the sources in these regime compare to BDWF time averaging without overlap while overlap sinc and J₀ time averaging attenuated the source in a variable integration time range compare to the boxcar (see black and red curve in Figure 35 and 37). The integration time range where the source is attenuated depends on the WF (sinc or J₀) and the overlap time bins which cause its to variate.

7.2.3 Time and frequency direction

Boxcar and BDWF averaging are processed across the time and the frequency direction. Each sky model is simulated with a JVLAC HR-MS of 300 timeslots and the integration time Δt^{hrms} is chosen within the range [0.5, 5.5]s at 1.4GHz. The simulation considered 150 channels of width $\Delta\nu^{hrms}$ chosen within the range [125., 750.]kHz. The result of the process was saved into a LR-MS of $100 \times \Delta t^{hrms}s$ synthesis, with $100 \times \Delta t^{hrms}$ integration time and 1 channel of width $50 \times \Delta\nu^{hrms}kHz$. We considered, $t_{start}^{hrms} = 0s$, $t_{start}^{lrms} = 100 \times \Delta t^{hrms}s$, $t_{end}^{hrms} = 300 \times \Delta t^{hrms}s$ and $t_{end}^{lrms} = 200 \times \Delta t^{hrms}s$, $\nu_{start}^{hrms} = \Delta\nu^{hrms}kHz$, $\nu_{start}^{lrms} = \nu_{start}^{hrms} \times 50kHz$, $\nu_{end}^{hrms} = \nu_{start}^{hrms} \times 150kHz$ and $\nu_{end}^{lrms} = \nu_{start}^{hrms} \times 100kHz$.

The source flux density as a function of the integration time and channels width is depicted in Figures 40 to 45. The study compared the maximal integration time and channels width of two dimensional boxcar averaging (Bx-avg- 0×0) with that of 2° FoV of two BDWF averaging (Bl-sinc- 100×50 , Bl-sinc- 100×50 , Bl-J₀- 0×50 , Bl-J₀- 100×50 , Bl-BW- 0×50 and Bl-BW- 100×50). The results show that:

- "Regime 1": BDWF averaging maintained the flux density of the source for large integration time and wide channels widths. However, it is seen on the Figures that the curve of sources in the regime are closer to the affine line $y=1$ when the integration time and channels width increases.
- "Regime 2" and "Regime 3": It appear that the curves of the overlaps sinc and J₀ are below the one of boxcar averaging for all integrations and channels widths. The results suggest that one do not need to worry about sidelobes confusion for large integration time and wide channels widths.

Discussion: The study describes the performances of the WF under consideration. As presented, the sinc performed well in *Regime 1* compared to the J₀ while the J₀ performed well in *Regime 2* and *Regime 3* compared to the sinc. The great potential of the two dimensional overlap sinc and the J₀ is the accurate performance in the three regimes. An interesting improvement of this study is the possibility of using WF capable of providing higher dynamic range images. This advanced techniques is required by the radio telescopes of the future.

8 REALISTIC FIELD SYNTHESIS: 3C147

9 CONCLUSIONS

The goal of this paper was threefold. The original motivation behind the work presented in this

paper was to **** windowing functions***

The second objective was to study ****first algorithm data compression***

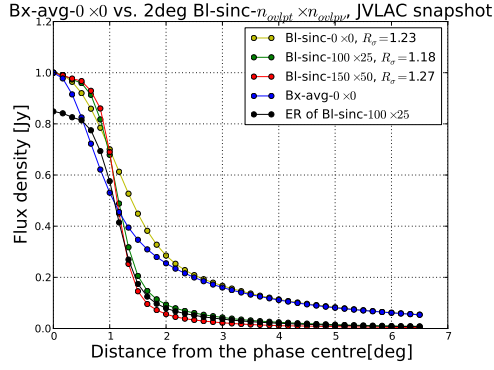


Figure 21. Time and frequency direction sinc filter applied on a 2° FoV JVLAC surveys observing a 1Jy source move from the phase centre for 150s integration synthesis at 6.25MHz bandwidth, natural weighting.

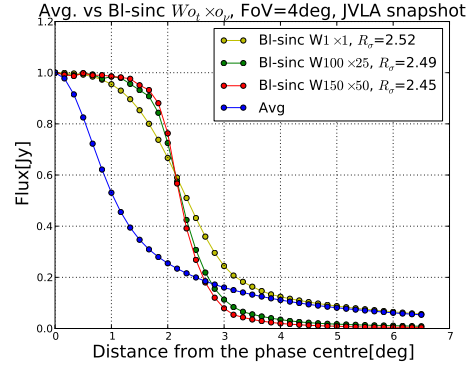


Figure 22. Time and frequency direction sinc filter applied on a 4° FoV JVLAC surveys observing a 1Jy source move from the phase centre for 150s integration synthesis at 6.25MHz bandwidth, natural weighting.

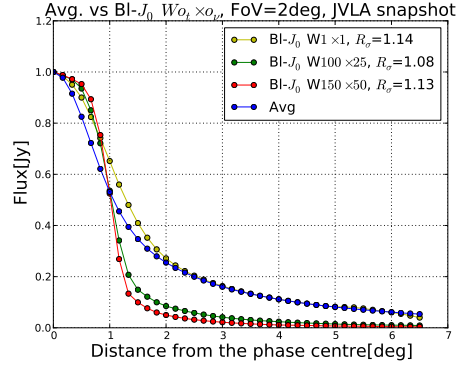


Figure 23. Time and frequency direction Bessel first kind filter applied on a 2° FoV JVLAC surveys observing a 1Jy source move from the phase centre for 150s integration synthesis at 6.25MHz bandwidth, natural weighting.

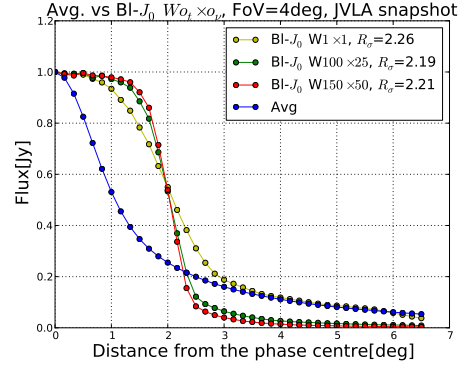


Figure 24. Time and frequency direction Bessel first kind filter applied on a 4° FoV JVLAC surveys observing a 1Jy source move from the phase centre for 150s integration synthesis at 6.25MHz bandwidth, natural weighting.

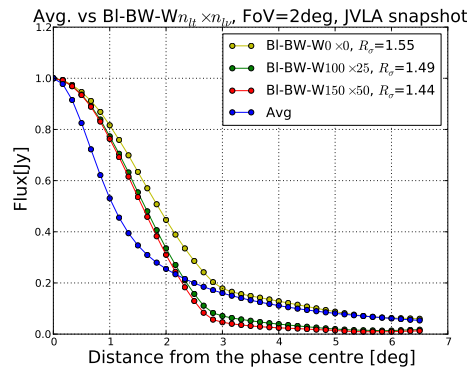


Figure 25. Time and frequency direction Butterworth filter applied on a 2° FoV JVLAC surveys observing a 1Jy source move from the phase centre for 150s integration synthesis at 6.25MHz bandwidth, natural weighting.

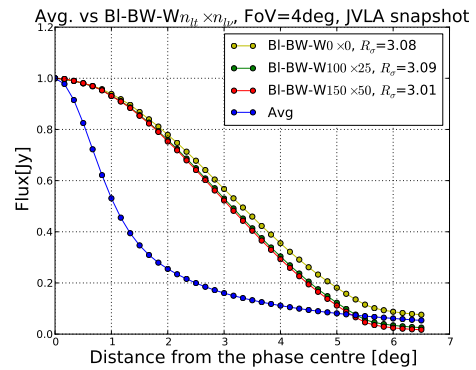


Figure 26. Time and frequency direction Butterworth filter applied on a 4° FoV JVLAC surveys observing a 1Jy source move from the phase centre for 150s integration synthesis at 6.25MHz bandwidth, natural weighting.

Figure 27. Time and frequency direction Butterworth filter applied on a 4° FoV JVLAC surveys observing a 1Jy source move from the phase centre for 150s integration synthesis at 6.25MHz bandwidth, natural weighting.

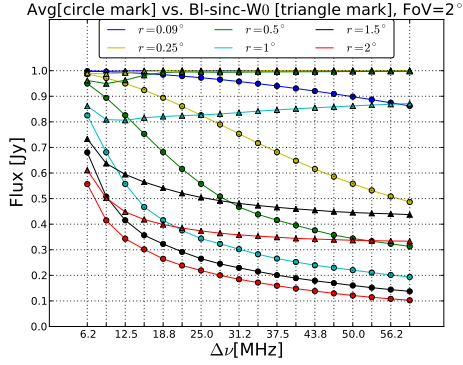


Figure 28. Response to a 1Jy source at different positions, as a function of bandwidth with 2° frequency sinc filter.

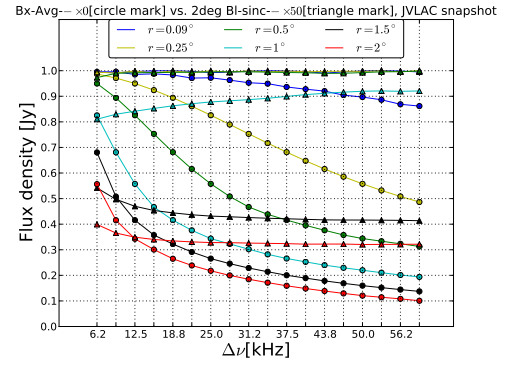


Figure 29. Response to a 1Jy source at different positions, as a function of bandwidth with 2° frequency overlap sinc filter.

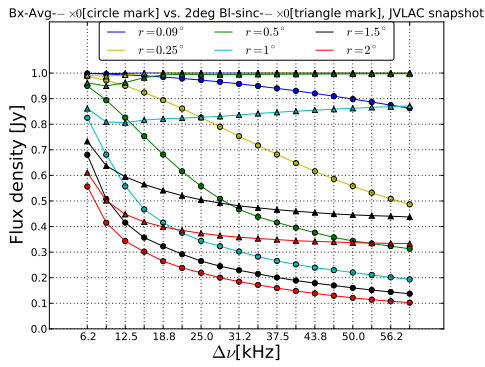


Figure 30. Response to a 1Jy source at different positions, as a function of bandwidth with 2° frequency Bessel first kind of order zero filter.

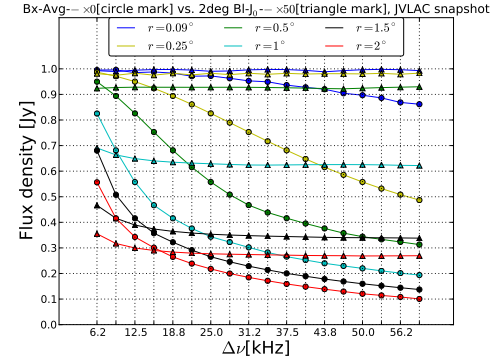


Figure 31. Response to a 1Jy source at different positions, as a function of bandwidth with 2° frequency overlap Bessel first kind filter of order zero.

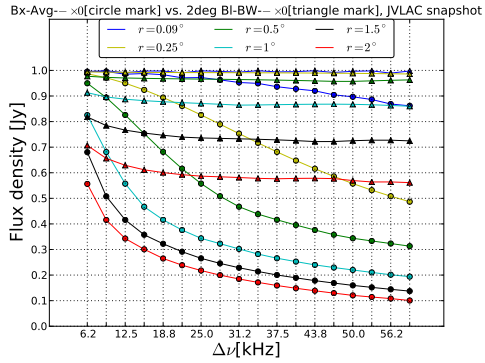


Figure 32. Response to a 1Jy source at different positions, as a function of bandwidth with 2° frequency Butterworth filter.

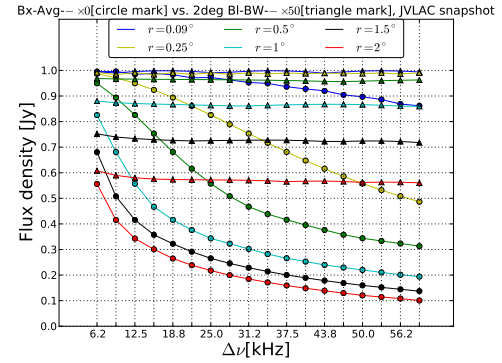


Figure 33. Response to a 1Jy source at different positions, as a function of bandwidth with 2° frequency overlap Butterworth filter.

The final objective was to ****second algorithm data compression and out field suppression***

Drawback and futures works*** drawback and futures works*****

ACKNOWLEDGEMENTS

The research made use of MeqTrees software system designed to implement numerical models for third-generation calibration (3GC) Python Extensions for Interferometry and Python. The research has been supported by the South Africa National Research Foundation.

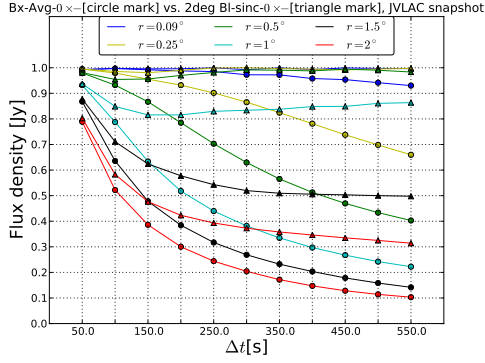


Figure 34. Response to a 1Jy source at different positions, as a function of integration time with 2° time sinc filter.

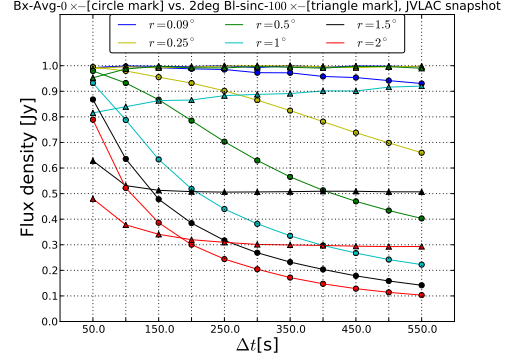


Figure 35. Response to a 1Jy source at different positions, as a function of integration time with 2° time overlap sinc filter.

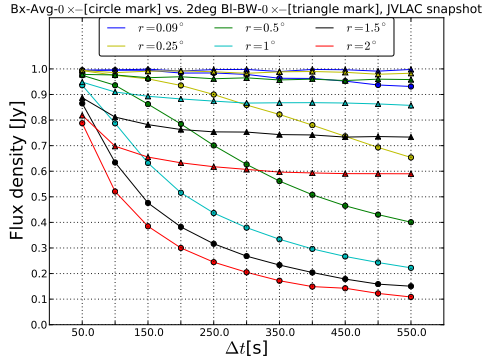


Figure 36. Response to a 1Jy source at different positions, as a function of integration time with 2° time Bessel first kind of order zero filter.

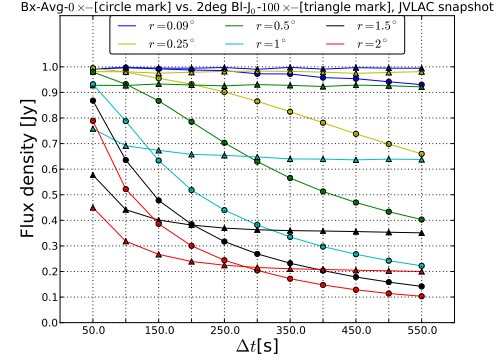


Figure 37. Response to a 1Jy source at different positions, as a function of integration time with 2° time overlap Bessel first kind of order zero filter.

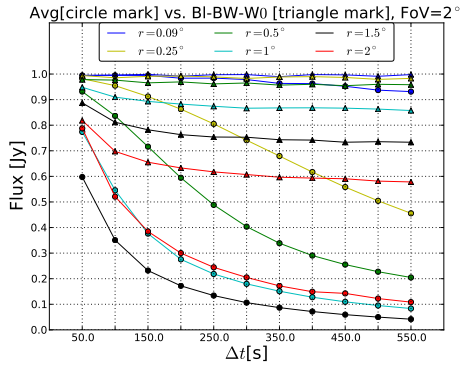


Figure 38. Response to a 1Jy source at different positions, as a function of integration time with 2° time Butterworth filter.

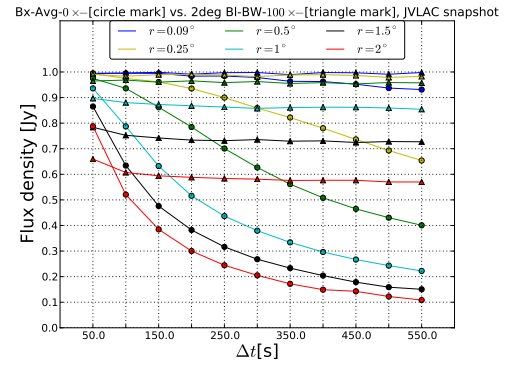


Figure 39. Response to a 1Jy source at different positions, as a function of integration time with 2° time overlap Butterworth filter.

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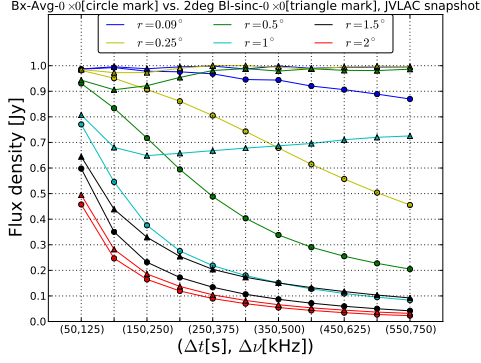


Figure 40. Response to a 1Jy source at different positions, as a function of integration time and bandwidth; with 2° frequency sinc filter.

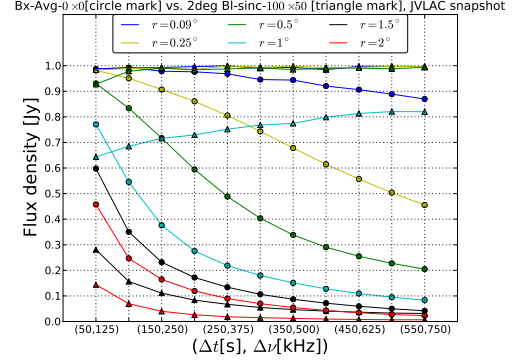


Figure 41. Response to a 1Jy source at different positions, as a function of integration time and bandwidth; with 2° frequency overlap sinc filter.

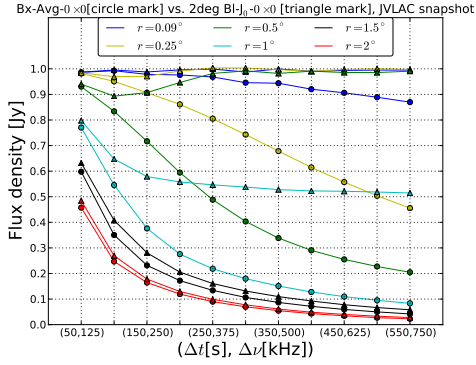


Figure 42. Response to a 1Jy source at different positions, as a function of integration time and bandwidth, with 2° frequency Bessel first kind of order zero filter.

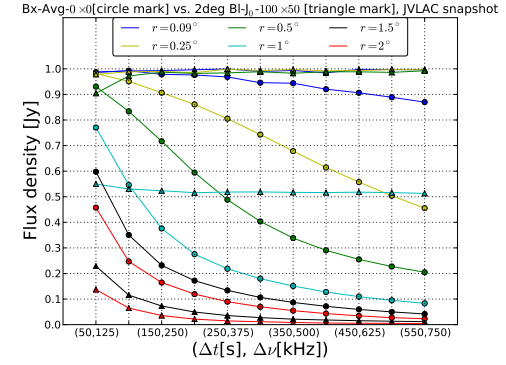


Figure 43. Response to a 1Jy source at different positions, as a function of integration time and bandwidth, with 2° frequency overlap Bessel first kind of order zero filter.

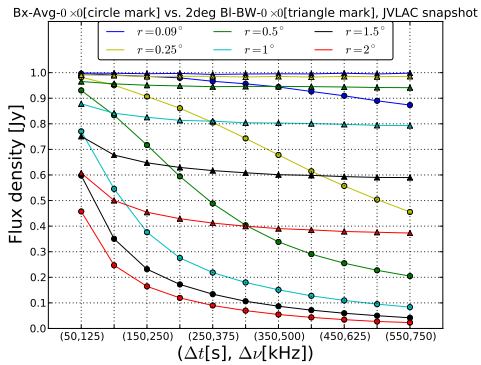


Figure 44. Response to a 1Jy source at different positions, as a function of integration time and bandwidth, with 2° frequency Butterworth filter.

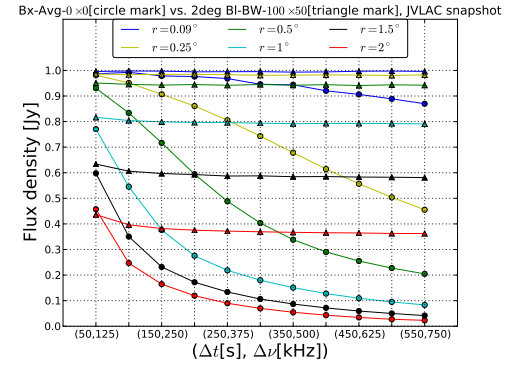


Figure 45. Response to a 1Jy source at different positions, as a function of integration time and bandwidth, with 2° frequency overlap Butterworth filter.

APPENDIX A: DERIVATION OF COMPLEX MATRICES

The complex matrices used in section 4 are explicitly derived in this appendix. In Eq.34, the matrices $\mathbf{C}_{(t,\nu)}^{block}$ and $\mathbf{W}_{pq,(t,\nu)}^{block}$ are blocks diagonals both of size $(4n_t n_\nu) \times (4n_t n_\nu)$ explicitly

expressed as follow:

$$\mathbf{C}_{(t,\nu)}^{block} = \begin{bmatrix} \mathbf{c}_{(t,\nu)} & 0 & 0 & 0 \\ 0 & \mathbf{c}_{(t,\nu)} & 0 & 0 \\ 0 & 0 & \mathbf{c}_{(t,\nu)} & 0 \\ 0 & 0 & 0 & \mathbf{c}_{(t,\nu)} \end{bmatrix}$$

$$\mathbf{W}_{pq,(t,\nu)}^{block} = \begin{bmatrix} \mathbf{W}_{pq,(t,\nu)} & 0 & 0 & 0 \\ 0 & \mathbf{W}_{pq,(t,\nu)} & 0 & 0 \\ 0 & 0 & \mathbf{W}_{pq,(t,\nu)} & 0 \\ 0 & 0 & 0 & \mathbf{W}_{pq,(t,\nu)} \end{bmatrix}$$

In Eq.35, the matrices $\mathbf{C}_{pq,(t,\nu)}^{block,nblock}$ and $\mathbf{W}_{pq,(t,\nu)}^{block,nblock}$ are blocks diagonals both of size $(4N_v^{pq}n_t n_\nu) \times (4N_v^{pq}n_t n_\nu)$, and the sampled visibilities $\mathbf{V}_{pq,(t,\nu)}^{samp,nblock}$ is a one row matrix of size $(N_v^{pq}4n_t n_\nu) \times (4n_t n_\nu)$ made of $\mathbf{V}_{pq,(t,\nu)}^{samp}$. These matrices are explicitly expressed as follow:

$$\mathbf{C}_{pq,(t,\nu)}^{block,nblock} = \begin{bmatrix} \mathbf{C}_{pq,(t,\nu)}^{block} & \dots & 0 & \dots & 0 \\ \vdots & \vdots & \vdots & \vdots & \vdots \\ 0 & \dots & \mathbf{C}_{pq,(t,\nu)}^{block} & \dots & 0 \\ \vdots & \vdots & \vdots & \vdots & \vdots \\ 0 & \dots & 0 & \dots & \mathbf{C}_{pq,(t,\nu)}^{block} \end{bmatrix}$$

$$\mathbf{W}_{pq,(t,\nu)}^{block,nblock} = \begin{bmatrix} \mathbf{W}_{pq,(t,\nu)}^{block} & \dots & 0 & \dots & 0 \\ \vdots & \vdots & \vdots & \vdots & \vdots \\ 0 & \dots & \mathbf{W}_{pq,(t,\nu)}^{block} & \dots & 0 \\ \vdots & \vdots & \vdots & \vdots & \vdots \\ 0 & \dots & 0 & \dots & \mathbf{W}_{pq,(t,\nu)}^{block} \end{bmatrix}$$

$$\mathbf{V}_{pq,(t,\nu)}^{samp,nblock} = \left(\mathbf{V}_{pq,(t,\nu)}^{samp,1}, \dots, \mathbf{V}_{pq,(t,\nu)}^{samp,k}, \dots, \mathbf{V}_{pq,(t,\nu)}^{samp,N_v^{pq}} \right)^\dagger.$$

In Eq.41, the matrix \mathbf{B} of size $(N_v 4n_t n_\nu) \times (4n_t n_\nu)$ is defined as follow:

$$\mathbf{B} = \begin{bmatrix} \mathbf{C}_{01,(t,\nu)}^{block,nblock} \cdot \mathbf{W}_{01,(t,\nu)}^{block,nblock} \\ \vdots \\ \mathbf{C}_{ik,(t,\nu)}^{block,nblock} \cdot \mathbf{W}_{ik,(t,\nu)}^{block,nblock} \\ \vdots \\ \mathbf{C}_{jl,(t,\nu)}^{block,nblock} \cdot \mathbf{W}_{jl,(t,\nu)}^{block,nblock} \end{bmatrix}$$

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